

Loops & Complexity
in
DIGITAL SYSTEMS

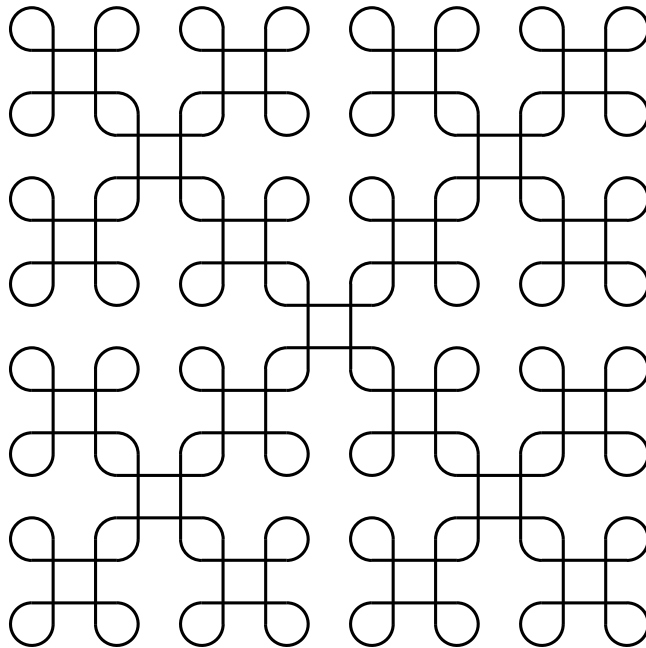
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Lecture Notes on Digital Electronics

Vol. 1

(work in endless progress)

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Introduction

Few legitimate questions about how to teach digital systems in *Ten Giga-Gate Per Chip Era* are waiting for an answer.

1. What means a *complex digital system*? How complex systems are designed using small and simple circuits?
2. How a digital system expands its size, increasing in the same time its speed? Are there simple mechanisms to be emphasized?
3. Is there a special mechanism allowing a “hierarchical growing” in a digital system? Or, how new features can be added in a digital system?

The *first question* occurs because already exist many different big systems which seem to have different degree of complexity. For example: big memory circuits and big processors. Both are implemented using a huge number of circuits, but the processors seem to be more “complicated” than the memories. In almost all text books complexity is related only with the dimension of the system. Complexity means currently only size, the concept being unable to make necessary distinctions in *Hundred Giga-Gate Per Chip Era*. The last improvements of the microelectronic technologies allow us to put on a Silicon die one hundred billions of gates, but the design tools are faced with more than the size of the system to be realized in this way. The *size* and the *complexity* of a digital system must be distinctly and carefully defined in order to have a more flexible conceptual environment for designing, implementing and testing systems in *Hundred Giga-Gate Per Chip Era*.

The *second question* rises in the same context of the big and the complex systems. Growing a digital system means both increasing its size and its complexity. How are correlated these two growing processes? The dynamic of *adding circuits* and of adding *adding features* seems to be very different and governed by distinct mechanisms.

The *third question* occurs in the hierarchical contexts in which the computation is defined. For example, Kleene’s functional hierarchy or Chomsky’s grammatical hierarchy are defined to explain how computation or formal languages used in computation evolve from simple to complex. Is this hierarchy reflected in a corresponding hierarchical organization of digital circuits? It is obvious that a sort of similar hierarchy must be hidden in the multitude of features already emphasized in the world of digital circuits. Let be the following list of usual terms: boolean functions, storing elements, automata circuits, finite automata, memory functions, processing functions, . . . , self-organizing processes, . . . Is it possible to disclose in this list a hierarchy, and more, is it possible to find similarities with previously exemplified hierarchies?

The first answer will be derived from the Kolmogorov-Chaitin *algorithmic complexity*: **the complexity of a circuit is related with the dimension of its shortest formal description**. A big circuit (a

circuit built using a big number of gates) can be simple or complex depending on the possibility to emphasize repetitive patterns in its structure. A no pattern circuit is a complex one because its description has the dimension proportional with its size. Indeed, for a complex, no pattern circuit each gate must be explicitly specified.

The second answer associates the **composition** with sizing and the **loop** with featuring. Composing circuits results in bigger structures with the same kind of functionality, while closing loops in a circuit new kind of behaviors are induced. Each new loop adds more *autonomy* to the system, because it increases the dependency of the output signals in the detriment of the input signals. Shortly, appropriate loops mean more autonomy that is equivalent sometimes with a new level of functionality.

The third answer is given by proposing *a taxonomy for digital systems based on the maximum number of included loops closed in a certain digital system*. The old distinction between combinational and sequential, applied only to **circuits**, is complemented with a classification taking into account the functional and structural diversity of the digital **systems** used in the contemporary designs. More, the resulting classification provides classes of circuits having direct correspondence with the levels belonging to Kleene's and Chomsky's hierarchies.

The first chapter: *What's a Digital System?* Few general questions are answered in this chapter. One refers to the position of digital system domain in the larger class of the sciences of computation. Another asks for presenting the ways we have to implement actual digital systems. The importance is also to present the correlated techniques allowing to finalize a digital product.

The second chapter: *Gates* The combinational circuits (0-OS) are introduced using a functional approach. We start with the simplest functions and, using different compositions, the basic simple functional modules are introduced. The distinction between simple and complex combinational circuits is emphasized, presenting specific techniques to deal with complexity.

The third chapter: *Memories* There are two ways to close a loop over the simplest functional combinational circuit: the *one-input decoder*. One of them offers the *stable structure* on which we ground the class of memory circuits (1-OS) containing: the elementary latches, the master-slave structures (the serial composition), the random access memory (the parallel composition) and the register (the serial-parallel composition). Few applications of storing circuits (pipeline connection, register file, content addressable memory, associative memory) are described.

The fourth chapter: *Automata* Automata (2-OS) are presented in the *fourth chapter*. Due to the second loop the circuit is able to evolve, more or less, autonomously in its own state space. This chapter begins presenting the simplest automata: the *T flip-flop* and the *JK flip-flop*. Continues with composed configurations of these simple structures: *counters* and related structures. Further, our approach makes distinction between the big sized, but simple *functional automata* (with the loop closed through a simple, recursive defined combinational circuit that can have any size) and the random, complex *finite automata* (with the loop closed through a random combinational circuit having the size in the same order with the size of its definition). The autonomy offered by the second loop is mainly used *to generate or to recognize specific sequences* of binary configurations.

The fifth chapter: *Processors* The circuits having three loops (3-OS) are introduced. The third loop may be closed in three ways: through a 0-OS, through an 1-OS or through a 2-OS, each of them being

meaningful in digital design. The first, because of the segregation process involved in designing automata using JK flip-flops or counters as state register. The size of the *random* combinational circuits that compute the state transition function is reduced, *in the most of case*, due to the increased autonomy of the device playing the role of the register. The second type of loop, through a memory circuit, is also useful because it increases the autonomy of the circuit so that the control exerted on it may be reduced (the circuit “knows more about itself”). The third type of loop, that interconnects two automata (an functional automaton and a control finite automaton), generates the most important digital circuits: the **processor**.

The sixth chapter: *Computing Machines* The effects of the fourth loop are shortly enumerated in the *sixth chapter*. The *computer* is the typical structure in 4-OS. It is also the support of the strongest segregation between the *simple* physical structure of the machine and the *complex* structure of the program (a symbolic structure). Starting from the fourth order the main functional up-dates are made structuring the symbolic structures instead of restructuring circuits. Few new loops are added in actual designs only for improving time or size performances, but not for adding new basic functional capabilities. For this reason our systematic investigation concerning the loop induced hierarchy stops with the fourth loop. The *toyMachine* behavioral description is revisited and substituted with a pure structural description.

The main stream of this book deals with the *simple* and the *complex* in digital systems, emphasizing them in the *segregation* process that opposes simple structures of circuits to the complex structures of symbols. The *functional information* offers the environment for segregating the simple circuits from the complex binary configurations.

When the simple is mixed up with the complex, the *apparent complexity* of the system increases over its *actual complexity*. We promote design methods which reduce the apparent complexity by segregating the simple from the complex. The best way to substitute the apparent complexity with the actual complexity is to drain out the chaos from order. One of the most important conclusions of this book is that the main role of the *loop* in digital systems is to *segregate* the *simple* from the *complex*, thus emphasizing and using the hidden resources of *autonomy*.

In the *digital systems domain* prevails the **art of disclosing the simplicity** because there exists the symbolic domain of functional information in which we may ostracize the complexity. But, the complexity of the process of disclosing the simplicity exhausts huge resources of imagination. This book offers only the starting point for the *architectural thinking*: the art of finding the right place of the interface between simple and complex in computing systems.

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Chapter 1

WHAT'S A DIGITAL SYSTEM?

Talking about Apple, Steve said, "The system is there is no system." Then he added, "that doesn't mean we don't have a process." Making the distinction between process and system allows for a certain amount of fluidity, spontaneity, and risk, while in the same time it acknowledges the importance of defined roles and discipline.

J. Young & W. Simon¹

A process is a strange mixture of rationally established rules, of imaginatively driven chaos, and of integrative mystery.

A possible good start in teaching about a complex domain is an *informal* one. The main problems are introduced friendly, using an easy approach. Then, little by little, a more rigorous style will be able to consolidate the knowledge and to offer formally grounded techniques. The digital domain will be disclosed here alternating informal "bird's-eye views" with simple, formalized real stuff. Rather than imperatively presenting the digital domain we intend to disclose it in small steps using a project oriented approach.

1.1 Framing the digital design domain

1.1.1 Digital Domain

In the electronic digital domain we work with two values only (see Figure 1.1):

¹They co-authored *iCon. Steve Jobs. The Greatest Second Act in the History of Business*, an unauthorized portrait of the co-founder of Apple.

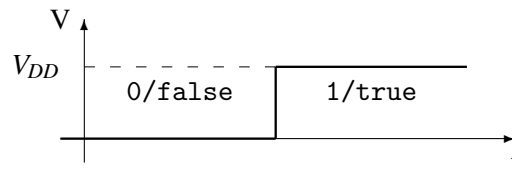


Figure 1.1: The two levels of the signal in the digital domain. Low level (0 Volt) for 0 or false, and high level (V_{DD} Volt) for 1 or true.

- 0, represented by the electrical value 0 V, having two meanings:
 - the numerical value 0
 - the logic value false
- 1, represented by the electrical value V_{DD} V, having two meanings:
 - the numerical value 1
 - the logic value true

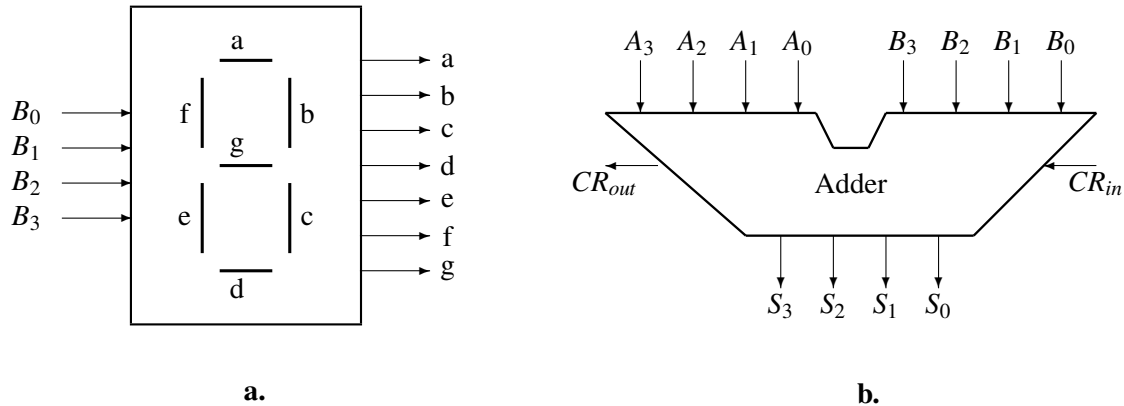


Figure 1.2: The version of digital circuits. **a.** Logic circuit: trans-coder for seven-segment display. **b.** Numeric circuit: four-bit numbers adder.

Consequently, there are two kinds of circuits (see Figure 1.2):

- logic circuits (Fig. 1.2a)
- numeric circuits (Fig. 1.2b)

Digital domain can be defined starting from two different, but complementary view points: the *structural* view point or the *functional* view point. The first version presents the digital domain as part of electronics, while the second version sees the digital domain as part of computer science.

1.1.2 Digital domain as part of electronics

Electronics started as a technical domain involved in processing continuously variable signals. Now the domain of electronics is divided in two sub-domains: *analogue electronics*, dealing with continuously variable signals and *digital electronics* based on elementary signals, called **bits**, which take only two different levels 0 and 1, but can be used to compose any complex signals. Indeed, a sequence of n bits is used to represent any number between 0 and $2^n - 1$, while a sequence of numbers can be used to approximate a continuously variable signal. Let us take first examples with 1-bit signals.

Example 1.1 *A disciplined driver starts the car's engine only if all four doors are closed and, in all occupied seats, the seat belts are connected. The key contact and the previous condition are the ones that start the engine. (This example is from [1].)*

The car is equipped with sensors for each door (d_1, d_2, d_3, d_4), for each seat (s_1, s_2, s_3, s_4), for each belt (b_1, b_2, b_3, b_4) and for the ignition key (k). The logic function that generates the start bit (s) is as follows:

$$\begin{aligned}
 s &= (\text{doors_are_closed}) \text{ AND } (\text{each_occupied_with_belt_on}) \text{ AND } (\text{key_is_on}) \\
 s &= (d_1 \text{ AND } d_2 \text{ AND } d_3 \text{ AND } d_4) \text{ AND} \\
 &\quad ((b_1 \text{ OR } (\text{NOT } b_1) \text{ AND } (\text{NOT } s_1)) \text{ AND} \\
 &\quad (b_2 \text{ OR } (\text{NOT } b_2) \text{ AND } (\text{NOT } s_2)) \text{ AND} \\
 &\quad (b_3 \text{ OR } (\text{NOT } b_3) \text{ AND } (\text{NOT } s_3)) \text{ AND} \\
 &\quad (b_4 \text{ OR } (\text{NOT } b_4) \text{ AND } (\text{NOT } s_4))) \text{ AND} \\
 &\quad k)
 \end{aligned}$$

In algebraic notation:

$$s = (d_1 \cdot d_2 \cdot d_3 \cdot d_4) \cdot ((b_1 + b_1' \cdot s_1') \cdot (b_2 + b_2' \cdot s_2') \cdot (b_3 + b_3' \cdot s_3') \cdot (b_4 + b_4' \cdot s_4')) \cdot k$$

Because the operator AND, “·”, is usually omitted:

$$s = d_1 d_2 d_3 d_4 (b_1 + b_1' s_1')(b_2 + b_2' s_2')(b_3 + b_3' s_3')(b_4 + b_4' s_4')k$$

The expression can be simplified because: $a + a'b = a + b$ (half-absorption rule).

Indeed, the car can start if each place has the belt on or is not occupied. Results the simplified form:

$$s = d_1 d_2 d_3 d_4 (b_1 + s_1')(b_2 + s_2')(b_3 + s_3')(b_4 + s_4')k$$

The Verilog description is:

```

module ignitionKey( output s ,
                    input  d1 , d2 , d3 , d4 , s1 , s2 , s3 , s4 ,
                    b1 , b2 , b3 , b4 , k);
    assign s = d1 & d2 & d3 & d4 & (b1 | ~s1) &
                    (b2 | ~s2) &
                    (b3 | ~s3) &
                    (b4 | ~s4) & k ;
endmodule

```

◇

Example 1.2 Let be the analogue, continuously variable, signal in Figure 1.3. It can be approximated by values sampled in discrete moments of time determined by the positive transitions of a square wave periodic signal called **clock**. It switches with a frequency of $1/T$. The value of the signal is measured in units u (for example, $u = 100\text{mV}$ or $u = 10\mu\text{A}$). The operation is called analog to digital conversion, and it is performed by an **analog to digital converter** – ADC. Results the following sequence of numbers:

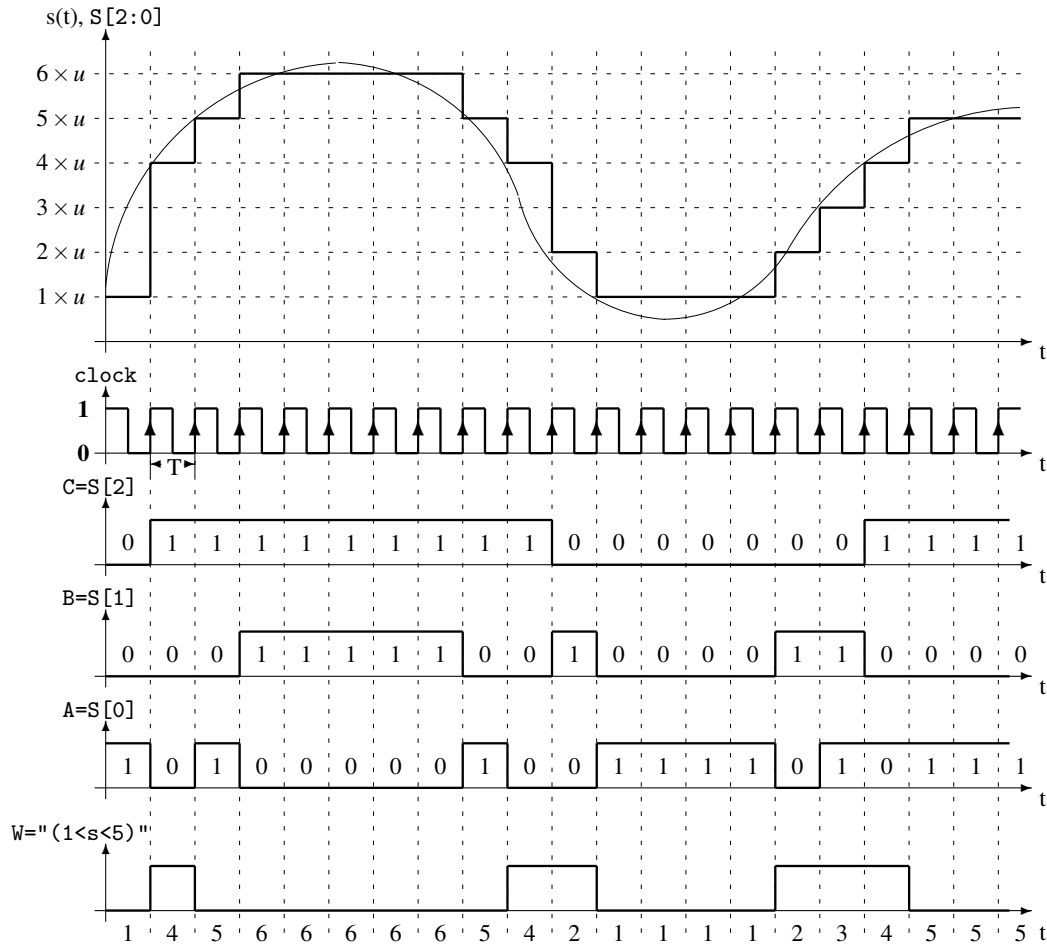


Figure 1.3: **Analogue to digital conversion.** The analog signal, $s(t)$, is sampled at each T using the unit measure u , and results the three-bit digital signal $S[2:0]$. **A first application:** the one-bit digital signal $W = "(1 < s < 5)"$ indicates, by its active value 1, the time interval when the digital signal is strictly included between $1u$ and $5u$. The three-bit result of conversion is $S[2:0]$.

$$\begin{aligned}
 s(0 \times T) &= 1 \text{ units} \Rightarrow 001, \\
 s(1 \times T) &= 4 \text{ units} \Rightarrow 100, \\
 s(2 \times T) &= 5 \text{ units} \Rightarrow 101, \\
 s(3 \times T) &= 6 \text{ units} \Rightarrow 110, \\
 s(4 \times T) &= 6 \text{ units} \Rightarrow 110, \\
 s(5 \times T) &= 6 \text{ units} \Rightarrow 110, \\
 s(6 \times T) &= 6 \text{ units} \Rightarrow 110,
 \end{aligned}$$

$s(7 \times T) = 6 \text{ units} \Rightarrow 110,$
 $s(8 \times T) = 5 \text{ units} \Rightarrow 101,$
 $s(9 \times T) = 4 \text{ units} \Rightarrow 100,$
 $s(10 \times T) = 2 \text{ units} \Rightarrow 010,$
 $s(11 \times T) = 1 \text{ units} \Rightarrow 001,$
 $s(12 \times T) = 1 \text{ units} \Rightarrow 001,$
 $s(13 \times T) = 1 \text{ units} \Rightarrow 001,$
 $s(14 \times T) = 1 \text{ units} \Rightarrow 001,$
 $s(15 \times T) = 2 \text{ units} \Rightarrow 010,$
 $s(16 \times T) = 3 \text{ units} \Rightarrow 011,$
 $s(17 \times T) = 4 \text{ units} \Rightarrow 100,$
 $s(18 \times T) = 5 \text{ units} \Rightarrow 101,$
 $s(19 \times T) = 5 \text{ units} \Rightarrow 101,$
 $s(20 \times T) = 5 \text{ units} \Rightarrow 101,$
 ...

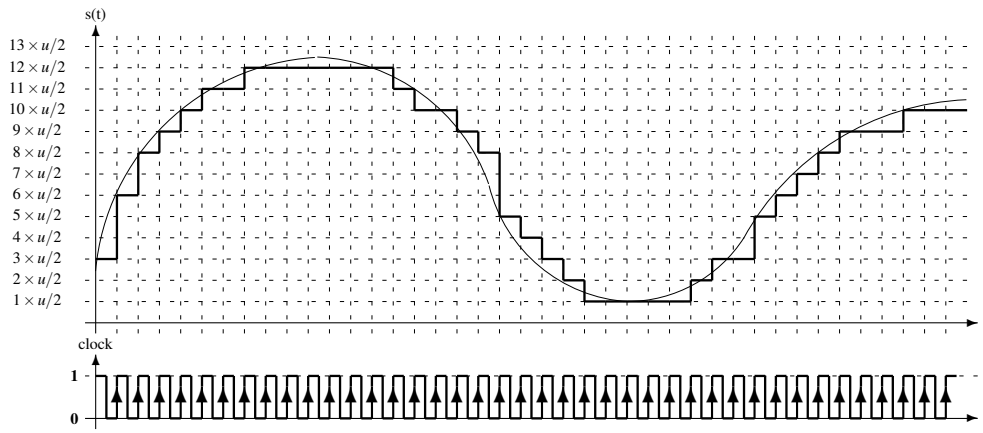


Figure 1.4: **More accurate analogue to digital.** The analog signal is sampled at each $T/2$ using the unit measure $u/2$.

If a more accurate representation is requested, then both, the sampling period, T and the measure units u must be reduced. For example, in Figure 1.4 both, T and u are halved. A better approximation is obtained with the price of increasing the number of bits used for representation. Each sample is represented on 4 bits instead of 3, and the number of samples is doubled. This second, more accurate, conversion provides the following stream of binary data:

<0011, 0110, 1000, 1001, 1010, 1011, 1011, 1100, 1100, 1100, 1100, 1100, 1100, 1100, 1011, 1010, 1010, 1001, 1000, 0101, 0100, 0011, 0010, 0001, 0001, 0001, 0001, 0001, 0010, 0011, 0011, 0101, 0110, 0111, 1000, 1001, 1001, 1001, 1010, 1010, 1010, ...>

◇

An ADC is characterized by two main parameters:

- the sampling rate: expressed in samples per second – SPS – or by the sampling frequency – $1/T$
- the resolution: the number of bits used to represent the value of a sample

Commercial ADC are provided with resolution in the range of 6 to 24 bits, and the sample rate exceeding 3 GSPS (giga SPS). At the highest sample rate the resolution is limited to 12 bits.

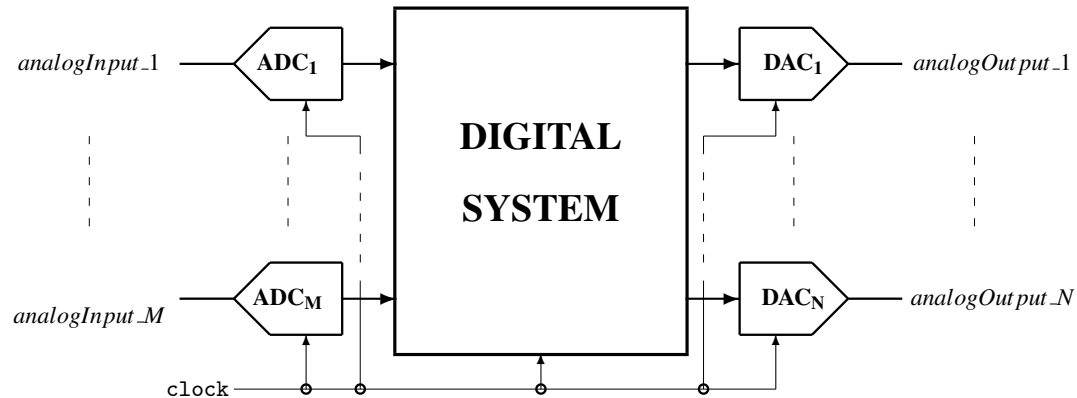


Figure 1.5: **Generic digital electronic system.**

The generic digital electronic system is represented in Figure 1.5, where:

- $analogInput_i$, for $i = 1, \dots, M$, provided by various sensors (microphones, ...), are sent to the input of M ADCs
- ADC_i converts $analogInput_i$ in a stream of binary coded numbers, using an appropriate sampling interval and an appropriate number of bits for approximating the level of the input signal
- DIGITAL SYSTEM processes the M input streams of data providing on its outputs N streams of data applied on the input of N Digital-to-Analog Converters (DAC)
- DAC_j converts its input binary stream to $analogOutput_j$
- $analogOutput_j$, for $j = 1, \dots, N$, are the outputs of the electronic system used to drive various actuators (loudspeakers, ...)
- $clock$ is the synchronizing signal applied to all the components of the system; it is used to trigger the moments when the signals are ready to be used and the subsystems are ready to use the signals.

While loosing something at conversion, we are able to gain at the level of processing. The good news is that the loosing process is under control, because both, the accuracy of conversion and of digital processing are highly controllable.

In this stage we are able to understand that the internal structure of DIGITAL SYSTEM from Figure 1.5 must have the possibility to do deal with **binary signals** which must be **stored & processed**. The signals are stored synchronized with the active edge of the **clock** signal, while for processing are used

circuits dealing with two distinct values: **0** and **1**. Usually, the value 0 is represented by the low voltage, currently 0, while the value 1 by high voltage, currently $\sim 1V$. Consequently, two distinct kinds of circuits can be emphasized in this stage:

- **registers**: used to *register*, synchronously with the active edge of the clock signal, the n -bit binary configuration applied on its inputs
- **logic circuits**: used to implement a correspondence between **all** the possible combinations of 0s and 1s applied on its m -bit input and the binary configurations generated on its n -bit output.

Example 1.3 *Let us consider a system with one analog input digitized with a low accuracy converter which provides only three bits (like in the example presented in Figure 1.3). The one-bit output, w , of the Boolean (logic) circuit² to be designed, let's call it `window`, must be active (on 1) each time when the result of conversion is less than 5 and greater than 1. In Figure 1.3 the wave form represents the signal w for the particular signal represented in the first wave form. The transfer function of the circuit is represented in the table from Figure 1.6a, where: for three binary input configurations, $S[2:0] = \{C,B,A\} = 010 \mid 011 \mid 100$, the output must take the value 1, while for the rest the output must be 0. Pseudo-formally, we write:*

$$W = 1 \text{ when } \begin{array}{l} ((\text{not } C = 1) \text{ and } (B = 1) \quad \text{and } (\text{not } A = 1)) \text{ or} \\ ((\text{not } C = 1) \text{ and } (B = 1) \quad \text{and } (A = 1)) \quad \text{or} \\ ((C = 1) \quad \text{and } (\text{not } B = 1) \text{ and } (\text{not } A = 1)) \end{array}$$

Using the Boolean logic notation:

$$W = C' \cdot B \cdot A' + C' \cdot B \cdot A + C \cdot B' \cdot A' = C'B(A' + A) + CB'A' = C'B + CB'A'$$

The resulting logic circuit is represented in Figure 1.6b, where:

- three NOT circuits are used for generating the negated values of the three input variables: C, B, A
- one 2-input AND circuit computes $C'B$
- one 3-input AND circuit computes $CB'A'$
- one 2-input OR circuit computes the final OR between the previous two functions.

The circuit is simulated and synthesized using its description in the hardware description language (HDL) Verilog, as follows:

²See details about Boolean logic in the appendix **Boolean Functions**.

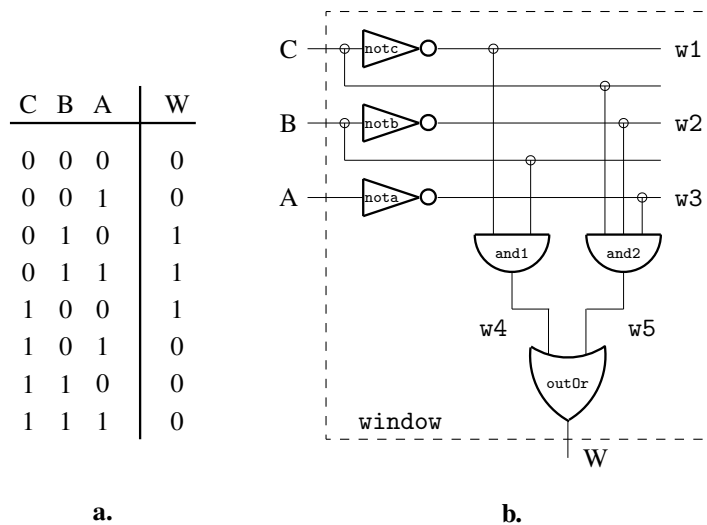


Figure 1.6: **The circuit window.** **a.** The truth table represents the behavior of the output for **all** binary configurations on the input. **b.** The circuit implementation.

```

/*****
File name:      window.v
Circuit name:   Window
Description:    the circuit detects the input in the range of (1,5)
*****/
module window( output W,
               input  C, B, A);

    wire      w1, w2, w3, w4, w5; // wires for internal connections

    not notc(w1, C),           // the instance 'notc' of the generic 'not'
        notb(w2, B),         // the instance 'notb' of the generic 'not'
        nota(w3, A);        // the instance 'nota' of the generic 'not'
    and and1(w4, w1, B),     // the instance 'and1' of the generic 'and'
        and2(w5, C, w2, w3); // the instance 'and2' of the generic 'and'
    or  outOr(W, w4, w5);   // the instance 'outOr' of the generic 'or'

endmodule

```

In Verilog, the entire circuit is considered a module, whose description starts with the keyword `module` and ends with the keyword `endmodule`, which contains:

- the declarations of two kinds of connections:
 - external connections associated to the name of the module as a list containing:
 - * the output connections (only one, W, in our example)
 - * the input connections (C, B and A)

- internal connections declared as `wire, w1, w2, ... w5`, used to interconnect the output of the internal circuits to the input of the internal circuits
- the instantiation of previously defined modules; in our example these are generic logic circuits expressed by keywords of the language, as follows:
 - circuits `not`, instantiated as `nota, notb, notc`; the first connection in the list of connections is the output, while the second is the input
 - circuits `and`, instantiated as `and1, and2`; the first connection in the list of connections is the output, while the next are the inputs
 - circuit `or`, instantiated as `outOr`; the first connection in the list of connections is the output, while the next are the inputs

The Verilog description is used for simulating and for synthesizing the circuit.

The simulation is done by instantiating the circuit `window` inside the simulation module `simWindow`:

```

/* *****
File name:      simWindow.v
Circuit name:   Simulation module for simWindow.v
Description:    generate stimulus for the module simWindow.v
***** */
  module simWindow;

    reg      A, B, C ;
    wire     W      ;

    initial begin
        {C, B, A} = 3'b000 ;
        #1 {C, B, A} = 3'b001 ;
        #1 {C, B, A} = 3'b010 ;
        #1 {C, B, A} = 3'b011 ;
        #1 {C, B, A} = 3'b100 ;
        #1 {C, B, A} = 3'b101 ;
        #1 {C, B, A} = 3'b110 ;
        #1 {C, B, A} = 3'b111 ;
        #1 $stop ;
    end

    window dut( W, C, B, A);

    initial $monitor( "S=%b_W=%b" ,
                     {C, B, A}, W);
  endmodule

```

◇

1.1.3 Modules in Verilog vs. Classes in Object Oriented Languages

What kind of language is the Verilog HDL? We will show it is a sort of Object Oriented Language. Let us design in Verilog a four-input adder modulo 2^8 .

```

/* *****
File: adder2.v
Describes: two-input mod256 adder
***** */
module adder2( output [7:0] out ,
               input  [7:0] in0 , in1 );

    assign out = in0 + in1 ;

endmodule

```

```

/* *****
File: adder4.v
Describes: four-input mod256 adder
***** */
module adder4( output [7:0] out ,
               input  [7:0] in0 , in1 , in2 , in3 );

    wire      [7:0] sum1 , sum2 ;

    adder2    add1(sum1 , in0 , in1)    ,
              add1(sum2 , in2 , in3)    ,
              add1(out , sum1 , sum2)   ;

endmodule

```

In C++ programming language the program for adding four numbers can be written using, instead of two modules, two classes, as follows:


```

/*****
File: adder2.cpp
Describes:
  - Constructor: describes a two-input integer adder
  - Methods: displays the behavior of adder2 for test
*****/
class adder2{public:
  int in1, in2, out;
  // Constructor
  adder2(int a, int b){
    in1 = a;
    in2 = b;
    out = in1 + in2;
  }
  // Method
  void displayAdd2(){
    cout << in1 << in2 << out << endl;
  }
};

```

```

/*****
File: adder4.cpp
Describes:
  - Constructor: describes a four-input integer adder
    + uses three instances of adder2: S1, S2, S3
  - Methods: displays the behavior of adder4 for test
*****/
class adder4{public:
  int in1, in2, in3, in4, out;
  // Constructor
  adder4(int a, int b, int c, int d){
    in1 = a;
    in2 = b;
    in3 = c;
    in4 = d;
    adder2 S1(a, b);
    adder2 S2(c, d);
    adder2 S3(S1.out, S2.out);
    out = S3.out;
  }
  // Method
  void displayAdd4(){
    cout << in1 << in2 << in3 << in4 << out << endl;
  }
};

```

The class `adder2` describe the two-input adder used to build, three times instantiated in class `adder4`,

a four input adder.

A class is more complex than a module because it can contain, as a method, the way the class is tested. In *Verilog* we have to define a distinct module, `testAdder2` or `testAdder4`, for simulation.

1.1.4 Digital domain as part of computer science

The domain of digital systems is considered, from the functional view point, as part of computing science. This, possible view point presents the digital systems as systems which *compute* their associated transfer functions. A digital system is seen as a sort of electronic system because of the technology used now to implement it. But, from a functional view point it is simply a computational system, because future technologies will impose maybe different physical ways to implement it (using, for example, different kinds of nano-technologies, bio-technologies, photon-based devices, ...). Therefore, we decided to start our approach using a functionally oriented introduction in digital systems, considered as a sub-domain of computing science. Technology dependent knowledge is always presented only as a supporting background for various design options.

Where can be framed the domain of digital systems in the larger context of computing science? A simple, informal definition of computing science offers the appropriate context for introducing digital systems.

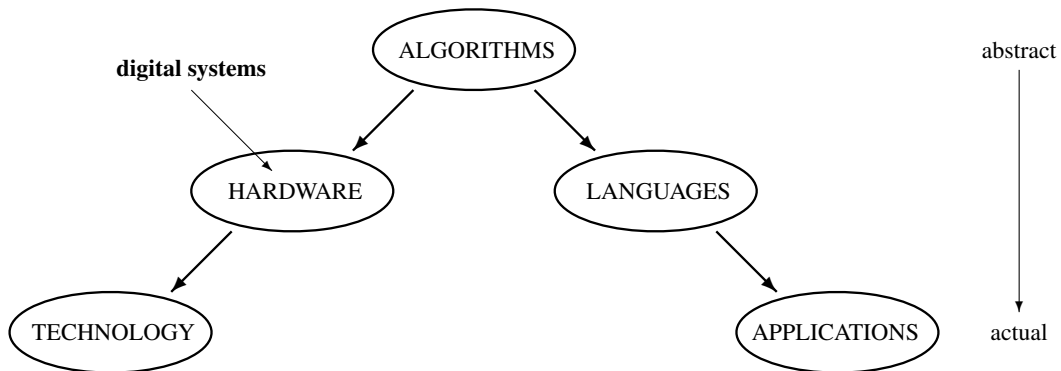


Figure 1.7: **What is computer science?** The domain of digital systems provides techniques for designing the hardware involved in computation.

Definition 1.1 *Computer science (see also Figure 1.7) means to study:*

- **algorithms,**
- *their **hardware** embodiment*
- *and their **linguistic** expression*

with extensions toward

- **hardware technologies**
- *and real **applications.** ◊*

The initial and the most *abstract level* of computation is represented by the algorithmic level. Algorithms specify *what* are the steps to be executed in order to perform a computation. The most *actual level* consists in two realms: (1) the huge and complex domain of the application software and (2) the very tangible domain of the real machines implemented in a certain technology. Both contribute to implement real functions (asked, or aggressively imposed, by the so called free market). An *intermediate level* provides the means to be used for allowing an algorithm to be embodied in a physical structure of a machine or in an informational structure of a program. It is about (1) the domain of the formal programming languages, and (2) the domain of hardware architecture. Both of them are described using specific and rigorous formal tools.

The hardware embodiment of computations is done in **digital systems**. What kind of formal tools are used to describe, in the most flexible and efficient way, a complex digital system? Figure 1.8 presents the formal context in which the description tools are considered. **Pseudo-code language** is an easy to understand and easy to use way to express algorithms. Anything about computation can be expressed using this kind of languages. By the rule, in a pseudo-code language we express, for our (human) mind, preliminary, not very well formally expressed, ideas about an algorithm. The “main user” of this kind of language is only the human mind. But, for building *complex* applications or for accessing advanced technologies involved in building *big* digital systems, we need refined, rigorous formal languages and specific styles to express computation. More, for a rigorous formal language we must take into account that the “main user” is a merciless machine, instead of a tolerant human mind. Elaborated **programming languages** (such as C++, Java, Prolog, Lisp) are needed for developing complex contexts for computation and to write using them real applications. Also, for complex hardware embodiments specific **hardware description languages**, HDL, (such as Verilog, VHDL, SystemC) are proposed.

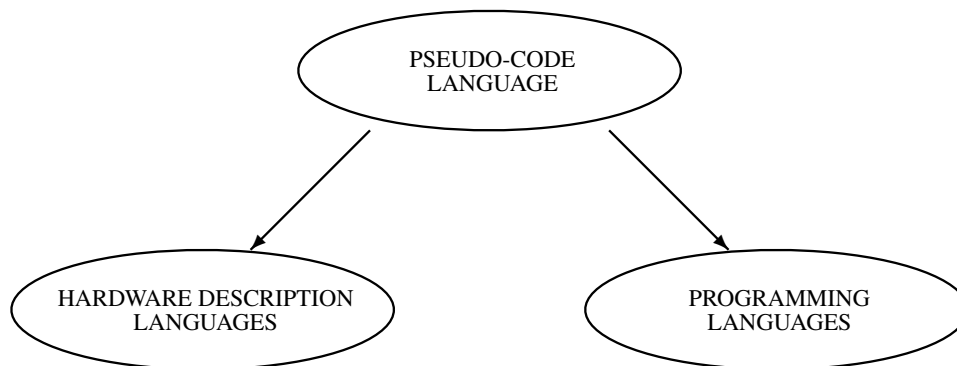


Figure 1.8: **The linguistic context in computer science.** Human mind uses pseudo-code languages to express informally a computation. To describe the circuit associated with the computation a rigorous HDL (hardware description language) is needed, and to describe the program executing the computation rigorous programming languages are used.

Both, general purpose programming languages and HDLs are designed to describe something for another program, mainly for a compiler. Therefore, they are more complex and rigorous than a simple pseudo-code language.

The starting point in designing a digital system is to describe it using what we call a **specification**, shortly, a **spec**. There are many ways to specify a digital system. In real life a hierarchy of specs are used, starting from high-level informal specs, and going down until the most detailed structural description is

provided. In fact, the design process can be seen as a stream of descriptions which starts from an idea about how the new object to be designed behaves, and continues with more detailed descriptions, in each stage more behavioral descriptions being converted in structural descriptions. At the end of the process a full structural description is provided. The design process is the long way from a spec about **what** we intend to do to another spec describing **how** our intention can be fulfilled.

At one end of this process there are innovative minds driven by the will to change the world. In these imaginative minds there is no knowledge about “*how*”, there is only willingness about “*what*”. At the other end of this process there are very skilled entities “*knowing*” *how* to do very efficiently what the last description provides. They do not care too much about the functionality they implement. Usually, they are machines driven by complex programs.

In between we need a mixture of skills provided by very well instructed and trained people. The role of the imagination and of the very specific knowledge are equally important.

How can be organized optimally a designing system to manage the huge complexity of this big chain, leading from an idea to a product? There is no system able to manage such a complex process. No one can teach us about how to organize a company to be successful in introducing, for example, a new processor on the real market. The real process of designing and imposing a new product is trans-systemic. It is a rationally adjusted chaotic process for which no formal rules can ever be provided.

Designing a digital system means to be involved in the middle of this complex process, usually far away from its ends. A **digital system designer** starts his involvement when the specs start to be almost rigorously defined, and ends his contribution before the technological borders are reached.

However, a digital designer is faced in his work with few levels of descriptions during the execution of a project. More, the number of descriptions increases with the complexity of the project. For a very simple project, it is enough to start from a spec and the structural description of the circuit can be immediately provided. But for a very complex project, the spec must be split in specs for sub-systems, each sub-system must be described first by its behavior. The process continues until enough simple sub-systems are defined. For them structural descriptions can be provided. The entire system is simulated and tested. If it works synthesizable descriptions are provided for each sub-system.

A good digital designer must be well trained in providing various descriptions using an HDL. She/he must have the ability to make, both behavioral and structural descriptions for circuits having any level of complexity. Playing with inspired partitioning of the system, a skilled designer is one who is able to use appropriate descriptions to manage the complexity of the design.

1.2 Defining a digital system

Digital systems belong to the wider class of the **discrete systems** (systems having a countable number of states). Therefore, a general definition for digital system can be done as a special case of discrete system.

Definition 1.2 A digital system, *DS*, in its most general form is defined by specifying the five components of the following quintuple:

$$DS = (X, Y, S, f, g)$$

where: $X \subseteq \{0, 1\}^n$ is the **input set** of n -bit binary configurations, $Y \subseteq \{0, 1\}^m$ is the **output set** of m -bit binary configurations, $S \subseteq \{0, 1\}^q$ is the **set of internal states** of q -bit binary configurations,

$$f : (X \times S) \rightarrow S$$

is the **state transition function**, and

$$g : (X \times S) \rightarrow Y$$

is the **output transition function**.

◇

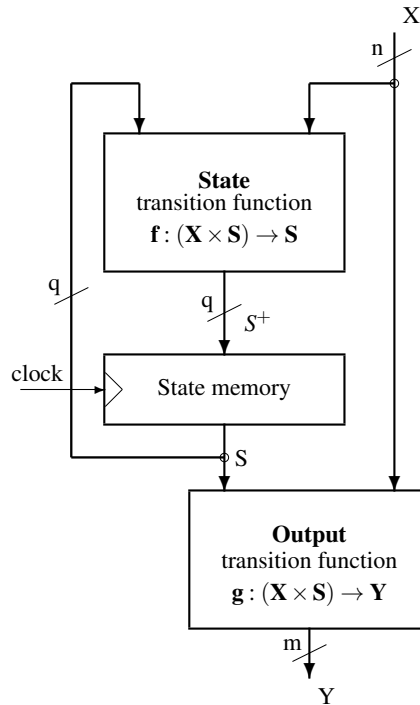


Figure 1.9: Digital system.

A digital system (see Figure 1.9) has two simultaneous evolutions:

- the evolution of its internal state which takes into account the current internal state and the current input, generating the next state of the system
- the evolution of its output, which takes into account the current internal state and the current input generating the current output.

The internal state of the system determines the partial autonomy of the system. The system behaves on its outputs taking into account both, the current input and the current internal state.

Because all the sets involved in the previous definition have the form $\{0, 1\}^b$, each of the b one-bit input, output, or state evolves in time switching between two values: 0 and 1. The previous definition specifies a system having a n -bit input, an m -bit output and a q -bit internal state. If $x_t \in X = \{0, 1\}^n$, $y_t \in Y = \{0, 1\}^m$, $s_t \in S = \{0, 1\}^q$ are values on input, output, and of state at the discrete moment of time t , then the behavior of the system is described by:

$$s_t = f(x_{t-1}, s_{t-1})$$

$$y_t = g(x_t, s_t)$$

While the current output is computed from the current input and the current state, the current state was computed using the previous input and the previous state. The two functions describing a discrete system belong to two distinct class of functions:

sequential functions : used to generate a sequence of values each of them iterated from its predecessor (an initial value is always provided, and the i -th value cannot be computed without computing all the previous $i - 1$ values); it is about functions such as $s_t = f(x_{t-1}, s_{t-1})$

non-sequential functions : used to compute an output value starting only from the current values applied on its inputs; it is about functions such as $y_t = g(x_t, s_t)$.

Depending on how the functions f and g are defined results a hierarchy of digital systems. More on this in the next chapters.

The variable **time** is essential for the formal definition of the sequential functions, but for the formal definition of the non-sequential ones it is meaningless. But, for the actual design of both, sequential and non-sequential function the time is a very important parameter.

Results the following requests for the *simplest embodiment* of an actual digital systems:

- the elements of the sets X , Y and S are binary cods of n , m and q bits – 0s and 1s – which are be coded by two electric levels; the current technologies work with 0 Volts for the value 0, and with a tension level in the range of 1-2 Volts for the value 1; thus, the system receives on its inputs:

$$X_{n-1}, X_{n-2}, \dots X_0$$

stores the internal state of form:

$$S_{q-1}, S_{q-2}, \dots S_0$$

and generate on its outputs:

$$Y_{m-1}, Y_{m-2}, \dots Y_0$$

where: $X_i, S_j, Y_k \in \{0, 1\}$.

- physical modules (see Figure 1.10), called **combinational logic circuits** – CLC –, to compute functions like $f(x_t, s_t)$ or $g(x_t, s_t)$, which *continuously follow*, by the evolution of their output values delayed with the *propagation time* t_p , any change on the inputs x_t and s_t (the shaded time interval on the wave out represent the transient value of the output)
- a “master of the discrete time” must be provided, in order to make consistent suggestions for the simple ideas as “previous”, “now”, “next”; it is about the special signal, already introduced, having form of a *square wave* periodic signal, with the period T which swings between the logic level 0 and the logic level 1; it is called cLock, and is used to “tick” the discrete time with its active edge (see Figure 1.11 where a clock signal, active on its positive edge, is shown)
- a storing support to memorize the state between two successive discrete moments of time is required; it is the **register** used to *register*, synchronized with the active edge of the clock signal, the state computed at the moment $t - 1$ in order to be used at the next moment, t , to compute a new state and a new output; the input must be stable a time interval t_{su} (*set-up* time) before the active edge of clock, and must stay unchanged t_h (*hold* time) after; the propagation time after the clock is t_p .

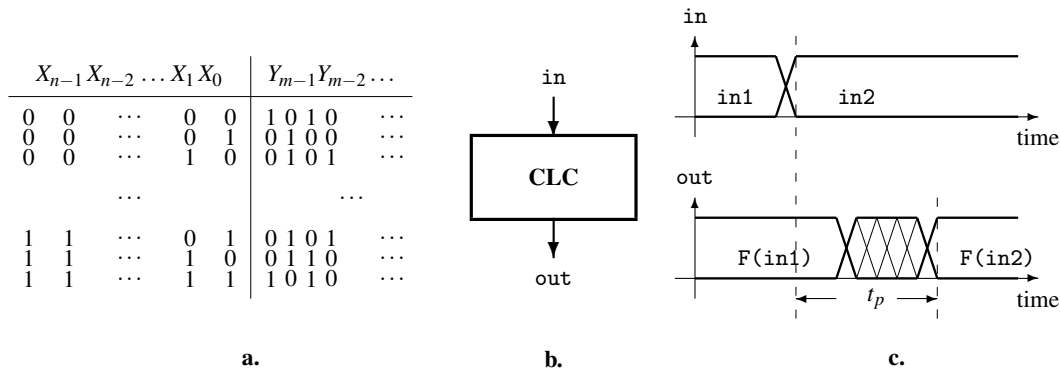


Figure 1.10: **The module for non-sequential functions.** **a.** The table used to define the function as a correspondence between **all** input binary configurations **in** and binary configurations **out**. **b.** The logic symbol for the *combinatorial logic circuit* – CLC – which computes $out = F(in)$. **c.** The wave forms describing the time behaviour of the circuit.

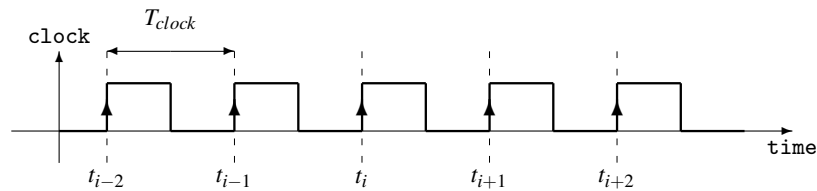


Figure 1.11: **The clock.** This clock signal is active on its positive edge (negative edge as active edge is also possible). The time interval between two positive transitions is the period T_{clock} of the clock signal. Each positive transition marks a discrete moment of time.

(More complex embodiment are introduced later in this text book. Then, the state will have a structure and the functional modules will result as multiple applications of this simple definition.)

The most complex part of defining a digital system is the description of the two functions f and g . The complexity of defining how the system behaves is managed by using various *Hardware Description Languages* – HDLs. The formal tool used in this text book is the **Verilog** HDL. The algebraic description of a digital system provided in Definition 1.2 will be expressed as the Verilog definition.

1.3 Different embodiment of digital systems

The physical embodiment of a digital system evolved, in the second part of the previous century, from circuits built using vacuum tubes to now a day complex systems implemented on a single die of silicon containing billions of components. We are here interested only by the actual stage of technology characterized by an evolutionary development and a possible revolutionary transition.

The evolutionary development is from the multi-chip systems approach to the *system on a chip* (SoC) implementations.

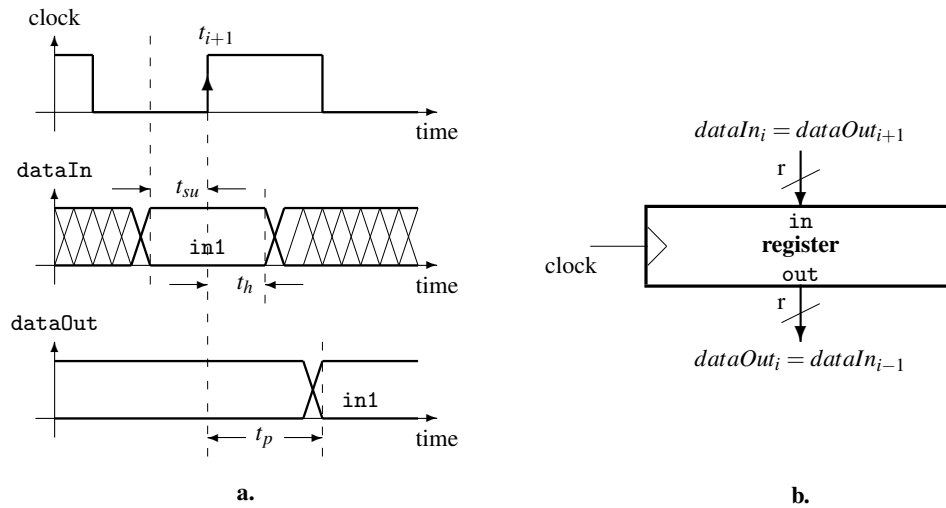


Figure 1.12: **The register.** **a.** The wave forms describing timing details about how the register switches around the active edge of clock. **b.** The logic symbol used to define the static behaviour of the register when both, inputs and outputs are stable between two active edges of the clock signal.

The revolutionary transition is from *Application Specific Integrated Circuit (ASIC)* approach to the **fully programmable solutions** for SoC.

SoC means integrating on a die a big system which, sometimes, involve more than one technology. Multi-chip approach was, and it is in many cases, necessary because of two reasons: (1) the big size of the system and, more important, (2) the need of use of few incompatible technologies. For example, there are big technological differences in implementing analog or digital circuits. If the circuit is analog, there is also a radio frequency sub-domain to be considered. The digital domain has also its specific sub-domain of the dynamic memories. Accommodating on the same silicon die different technologies is possible but the price is sometimes too big. The good news is that there are continuous technological developments providing cheap solutions for integrating previously incompatible technologies.

An ASIC provides very efficient solutions for well defined functions and for big markets. The main concern with this approach is the lack of functional flexibility on a very fast evolving market. Another problem with the ASIC approach is related with the “reusability” of the silicon area which is a very expensive resource in a digital system. For example, if the multiplication function is used in few stages of the algorithm performed by an ASIC, then a multiplication circuit must be designed and placed on silicon few times even if the circuits stay some- or many-times unused. An alternative solution provides only one multiplier which is “shared” by different stages of the algorithm, if possible.

There are different types of “programmable” digital systems:

- **reconfigurable systems:** are physical structures, having a set of useful features, can be configured, to perform a specific function, by the binary content of some specific storage registers called *configuring registers*; the flexibility of this approach is limited to the targeted application domain
- **programmable circuits:** are general purpose structures whose interconnection and simple functionality are both programmed providing any big and complex systems; but, once the functionality

in place, the system performs a fix function

- **programmable systems:** are designed using one or many programmable computing machines able to provide any transfer function between its inputs and outputs.

All these solutions must be evaluated taking into account their flexibility, speed performance, complexity, power consumption, and price. The *flexibility* is minimal for configurable systems and maximal for programmable circuits. *Speed performance* is easiest to be obtained with reconfigurable systems, while the programmable circuits are the laziest at big complexities. *Complexity* is maximal for programmable circuits and limited for reconfigurable systems. *Power consumption* is minimal for reconfigurable solutions, and maximal for programmable circuits. *Price* is minimal for reconfigurable systems, and maximal for programmable circuits. In all the previous evaluations programmable systems are avoided. Maybe this is the reason for which they provide overall the best solution!

Designing digital circuits is about the hardware support of programmable systems. This book provides knowledge on circuits, but the final target is to teach how to build various programmable structures. Optimizing a digital system means to have a good balance between the physical structure of circuits and the informational structure of programs running on them. Because the future of complex systems belongs to the programmable systems, the hardware support offered by circuits must be oriented toward programmable structures, whose functionality is actualized by the embedded information (program).

Focusing on programmable structures does not mean we ignore the skills involved in designing ASICs or reconfigurable systems. All we discuss about programmable structures applies also to any kind of digital structure. What will happen will be that at a certain level in the development of digital systems features for accepting program control will be added.

1.4 Correlated domains

Digital design must be preceded and followed by other disciplines. There are various prerequisites for attending a digital design course. These disciplines are requested for two reasons:

- the student must be *prepared* with an appropriate pool of knowledge
- the student must be *motivated* to acquire a new skill.

In an ideal world, a student is prepared to attend digital design classes by having knowledge about: *Boolean algebra* (logic functions, canonic forms, minimizing logic expressions), *Automata theory* (formal languages, finite automata, ... Turing Machine), *Electronic devices* (MOS transistor, switching theory), *Switching circuits* (CMOS structure, basic gates, transmission gate, static & dynamic behavior of the basic structures).

In the same ideal world, a student can be motivated to approach the digital design domain if he payed attention to *Theory of computation*, *Microprocessor architecture*, *Assembly languages*.

Attending the classes of Digital Systems is only a very important step on a long journey which suppose to attend a lot of other equally important disciplines. The most important are listed bellow.

Verification & testing For complex digital system verification and testing become very important tasks. The design must be verified to be sure that the intended functionality is in place. Then in each stage, on the way from the initial design to the fabrication of the actual chip, various tests are performed. Specific techniques are developed for verification and testing depending on the complexity of the design.

Specific design techniques are used to increase the efficiency of testing. *Design for testability* is a well developed sub-domain which helps us with design tricks for increasing the accuracy and speed of testing.

Physical design The digital system designer provides only a description. It is a program written in a HDL. This description must be used to build accurately an actual chip containing many hundred of million of circuits. It is a multi-stage process where after circuit design, simulation, synthesis, and functional verification, done by the digital design team, follow **layout design & verification, mask preparation, wafer fabrication, die test**. During this long process a lot of additional technical problem must be solved. A partial enumeration of them follows.

- **Clock distribution:** The *clock* signal is a pulse signal distributed almost uniformly on the whole area of the chip. For a big circuit the clock distribution is a critical problem because of the power involved and because of the accuracy of the temporal relation imposed for it.
- **Signal propagation:** Besides clock there are a lot of other signals which can be critical if they spread on big parts of the circuit area. The relation between these signals makes the problem harder.
- **Chip interface circuits:** The electrical charge of an interface circuit is much bigger than for the internal one. The capacitance load on pins being hundred times bigger the usual internal load, the output current for pin driver must be correspondingly.
- **Powering:** The switching energy is provided from a DC power supply. The main problem is to have enough energy right in time at the power connections of each circuit form the chip. Power distribution is made difficult by the inductive effect of the power connections.
- **Cooling:** The electrical energy introduced in circuit, through the power system, must be then, unfortunately, extracted as caloric energy (heat) by cooling it.
- **Packaging:** The silicon die is mounted in a package which must fulfil a lot of criteria. It must allow powering and cooling the die it contains. Also, it must provide hundreds or even thousands external connections. Not to mention protection to cosmic rays, ...
- **Board design:** The chips are designed to be mounted on boards where they are interconnected with other electronic components. Because of the very high density of connections, designing a board is a very complex job involving knowledge from a lot of related domains (electromagnetism, mechanics, chemistry, ...).
- **System design:** Actual applications are finalized as packaged systems containing one or many boards, sometimes interconnected with electro-mechanical devices. Putting together many components, powering them, cooling them, protecting them from disturbing external (electromagnetic, chemical, mechanical, ...) factors, adding esthetic qualities require multi-disciplinary skills.

For all these problems specific knowledge must be acquired attending special classes, course modules, or full courses.

Computer architecture Architectural thinking is a major tendency in the contemporary world. It is a way to discuss about the functionality of an object ignoring its future actual implementation. The architectural approach helps us to clarify first what we intend to build, unrestricted by the implementation issues. Computer architecture is a very important sub-domain of computer science. It allow us to develop independently the hardware domain and the software domain maintaining in the same time a high “communicating channel” between the two technologies: one referring to the physical structures and another involving the informational structure of programs.

Embedded systems In an advanced stage of development of digital system the physical structure of the circuits start to be interleaved with the informational structure of programs. Thus, the *functional* flexibility of the system and its efficiency is maximized. A digital system tend to be more and more a computational system. The computation become embedded into the core of a digital system. The discipline of embedded system or embedded computation³ starts to be a *finis coronat opus* of digital domain.

Project management Digital systems are complex systems. In order to finalize a real product a lot of activities must be correlated. Therefore, an efficient management is mandatory for a successful project. More, the management of the digital system project has some specific aspects to be taken into account.

Business & Marketing & Sales Digital systems are produced to be useful. Then, they must spread in our human community in the most appropriate way. Additional, but very related skills are needed to enforce on the market a new digital system. The knowledge about business, about marketing and sales is crucial for imposing a new design. A good, even revolutionary idea is necessary, but absolutely insufficient. The pure technical skills must be complemented by skills helping the access on the market, the only place where a design receives authentic recognition.

1.5 Problems

Problem 1.1 Let be the full 4-bit adder described in the following Verilog module:

```

module fullAdder (      output [3:0]    out      ,
                       output      crOut   , // carry output
                       input  [3:0]    in0    ,
                       input  [3:0]    in1    ,
                       input      crIn    ); // carry input

  wire    [4:0]    sum ;

  assign sum      = in0 + in1 + crIn ;
  assign out      = sum [3:0] ;
  assign crOut    = sum [4] ;

endmodule

```

³In DCAE chair of the Electronics Faculty, in Politehnica University of Bucharest this topics is taught as *Functional Electronics*, a course introduced in late 70s by the Professor Mihai Dr’ag’anescu.

Use the module `fullAdder` to design the following 16-bit full adder:

```

module bigAdder(output [15:0] out ,
                output crOut , // carry output
                input [15:0] in0 ,
                input [15:0] in1 ,
                input crIn ); // carry input

    // ???

endmodule

```

The resulting project will be simulated designing the appropriate test module.

Problem 1.2 Draw the block schematic of the following design:

```

module topModule( output [7:0] out ,
                  input [7:0] in1 ,
                  input [7:0] in2 ,
                  input [7:0] in3 );
    wire [7:0] wire1 , wire2 ;

    bottomModule mod1( .out(wire1 ) ,
                      .in1(in1 ) ,
                      .in2(in2 )) ,
                  mod2( .out(wire2 ) ,
                      .in1(wire1 ) ,
                      .in2(in3 )) ,
                  mod3( .out(out ) ,
                      .in1(in3 ) ,
                      .in2(wire2 )) ;

endmodule

```

```

module bottomModule(output [7:0] out ,
                    input [7:0] in1 ,
                    input [7:0] in2 );

    // ...

endmodule

```

Synthesize it to test your solution.

Problem 1.3 Let be the schematic representation of the design `topSyst` in Figure 1.13. Write the Verilog description of what is described in Figure 1.13. Test the result by synthesizing it.

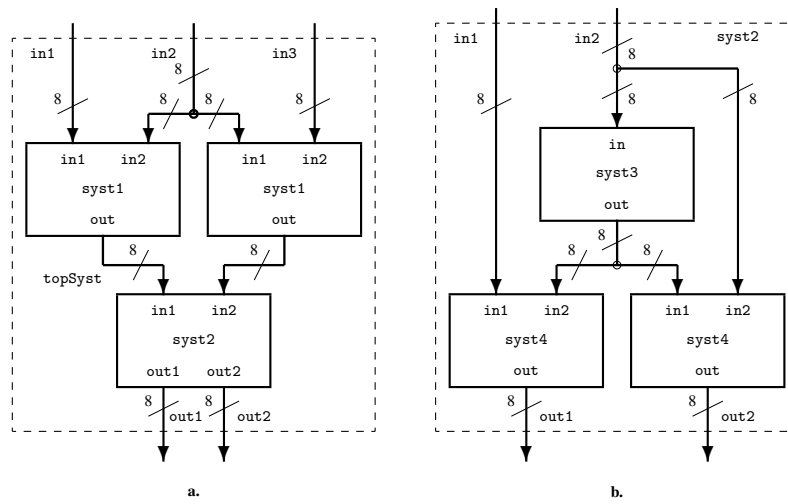


Figure 1.13: **The schematic of the design topSystem. a.** The top module `topSystem` **b.** The structure of the module `syst2`.

Chapter 2

GATES:

Zero order, no-loop digital systems

Belief #5: That qualitative as well as quantitative aspects of information systems will be accelerated by Moore's Law. ... *In the minds of some of my colleagues, all you have to do is identify one layer in a cybernetic system that's capable of fast change and then wait for Moore's Law to work its magic.*

Jaron Lanier¹

The Moore's Law applies to size not to complexity.

In this chapter we will forget for the moment about loops. Composition is the only mechanism involved in building a combinational digital system. No-loop circuits generate the class of history free digital systems whose outputs depend only by the current input variables, and are reassigned “continuously” at each change of inputs. Anytime the output results as a specific “combination” of inputs. No autonomy in combinational circuits, whose outputs obey “not to say a word” to inputs.

The combinational functions with n 1-bit inputs and m 1-bit outputs are called Boolean function and they have the following form:

$$f : \{0, 1\}^n \rightarrow \{0, 1\}^m.$$

For $n = 1$ only the NOT function is meaningful in the set of the 4 one-input Boolean functions. For $n = 2$ from the set of 16 different functions only few functions are currently used: AND, OR, XOR, NAND, NOR, NXOR. Starting with $n = 3$ the functions are defined only by composing 2-input functions. (For a short refresh see Appendix *Boolean functions*.)

Composing small gates results big systems. The growing process was governed in the last 40 years by Moore's Law². For a few more decades maybe the same growing law will act. But, starting from millions of gates per chip, it is very important what kind of circuits grow exponentially!

¹Jaron Lanier coined the term *virtual reality*. He is a computer scientist and a musician.

²The Moore's Law says the physical performances in microelectronics improve exponentially in time.

Composing gates results two kinds of big circuits. Some of them are structured following some *repetitive patterns*, thus providing simple circuits. Others grow *patternless*, providing complex circuits.

2.1 Simple, Recursive Defined Circuits

The first circuits used by designers were small **and** simple. When they were grew a little they were called big **or** complex. But, now when they are huge we must talk, more carefully, about *big sized simple circuits* **or** about *big sized complex circuits*. In this section we will talk about simple circuits which can be actualized at any size, i.e., their definitions don't depend by the number, n , of their inputs.

In the class of n -inputs circuits there are 2^{2^n} distinct circuits. From this tremendous huge number of logical function we use currently an insignificant small number of simple functions. What is strange is that these functions are sufficient for almost all the problem which we are confronted (or we are limited to be confronted).

One fact is clear: we can not design very big complex circuits because we can not specify them. The complexity must get away in another place (we will see that this place is the world of symbols). If we need big circuit they must remain simple.

In this section we deal with simple, if needed big, circuits and in the next with the complex circuits, but only with ones having small size.

From the class of the simple circuits we will present only some very usual such as *decoders*, *demultiplexors*, *multiplexors*, *adders* and *arithmetic-logic units*. There are many other interesting and useful functions. Many of them are proposed as problems at the end of this chapter.

2.1.1 Decoders

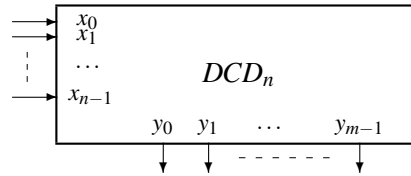
The simplest problem to be solved with a *combinational logic circuit* (CLC) is to answer the question: “*what is the value applied to the input of this one-input circuit?*”. The circuit which solves this problem is an **elementary decoder** (EDCD). It is a *decoder* because decodes its one-bit input value by activating distinct outputs for the two possible input values. It is *elementary* because does this for the smallest input word: the one-bit word. By decoding, the value applied to the input of the circuit is emphasized activating distinct signals (like lighting only one of n bulbs). This is one of the main functions in a digital system. Before generating an answer to the applied signal, the circuit must “know” what signal arrived on its inputs.

Informal definition

The n -input decoder circuit – DCD_n – (see Figure 2.1) performs one of the basic function in digital systems: with one of its m one-bit outputs specifies the binary configuration applied on its inputs. The binary number applied on the inputs of DCD_n takes values in the set $X = \{0, 1, \dots, 2^n - 1\}$. For each of these values there is one output – y_0, y_1, \dots, y_{m-1} – which is activated on 1 if its index corresponds with the current input value. If, for example, the input of a DCD_4 takes value 1010, then $y_{10} = 1$ and the rest 15 one-bit outputs take the value 0.

Formal definition

In order to rigorously describe and to synthesize a decoder circuit a formal definition is requested. Using *Verilog* HDL, such a definition is very compact certifying the non-complexity of this circuit.

Figure 2.1: The n -input decoder (DCD_n).

Definition 2.1 DCD_n is a combinational circuit with the n -bit input X , x_{n-1}, \dots, x_0 , and the m -bit output Y , y_{m-1}, \dots, y_0 , where: $m = 2^n$, with the behavioral Verilog description:

```

/* *****
File name:      dec.v
Circuit name:   Decoder
Description:    behavioral description of a n-input decoder
***** */
module dec #(parameter inDim = n)(input  [inDim - 1:0]      sel ,
                                output [(1 << inDim) - 1:0] out);
    assign out = 1 << sel;
endmodule

```

◇

The previous *Verilog* description is synthesisable by the current software tools which provide an efficient solution. It happens because this function is simple and it is frequently used in designing digital systems.

Recursive definition

The decoder circuit DCD_n for any n can be defined recursively in two steps:

- defining the elementary decoder circuit ($EDCD = DCD_1$) as the smallest circuit performing the decode function
- applying the *divide & impera* rule in order to provide the DCD_n circuit using $DCD_{n/2}$ circuits.

For the first step $EDCD$ is defined as one of the simplest and smallest logical circuits. Two one-input logical function are used to perform the decoding. Indeed, *parallel composing* (see Figure 2.2a) the circuits performing the simplest functions: $f_2^1(x_0) = y_1 = x_0$ (identity function) and $f_1^1(x_0) = y_0 = x_0'$ (NOT function), we obtain an ($EDCD$). If the output y_0 is active, it means the input is zero. If the output y_1 is active, then the input has the value 1.

In order to isolate the output from the input the *buffered EDCCD* version is considered *serial composing* an additional inverter with the previous circuit (see Figure 2.2b). Hence, the *fan-out* of $EDCCD$ does not depend on the fan-out of the circuit that drives the input.

The second step is to answer the question about how can be build a (DCD_n) for decoding an n -bit input word.

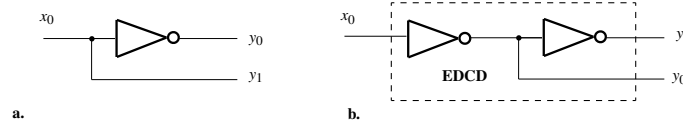


Figure 2.2: **The elementary decoder (EDCD).** **a.** The basic circuit. **b.** Buffered EDCD, a serial-parallel composition.

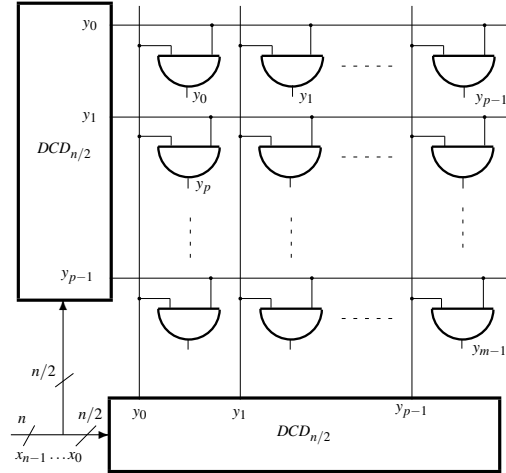


Figure 2.3: **The recursive definition of n -inputs decoder (DCD_n).** Two $DCD_{n/2}$ are used to drive a two dimension array of AND_2 gates. The same rule is applied for the two $DCD_{n/2}$, and so on until $DCD_1 = EDCD$ is needed.

Definition 2.2 The structure of DCD_n is recursive defined by the rule represented in Figure 2.3. The DCD_1 is an EDCD (see Figure 2.2b). \diamond

The previous definition is a constructive one, because provide an algorithm to construct a decoder for any n . It falls into the class of the “divide & impera” algorithms which reduce the solution of the problem for n to the solution of the same problem for $n/2$.

The quantitative evaluation of DCD_n offers the following results:

Size: $GS_{DCD}(n) = 2^n GS_{AND}(2) + 2GS_{DCD}(n/2) = 2(2^n + GS_{DCD}(n/2))$

$$GS_{DCD}(1) = GS_{EDCD} = 2$$

$$GS_{DCD}(n) \in O(2^n)$$

Depth: $D_{DCD}(n) = D_{AND}(2) + D_{DCD}(n/2) = 1 + D_{DCD}(n/2) \in O(\log n)$

$$D_{DCD}(1) = D_{EDCD} = 2$$

Complexity: $C_{DCD} \in O(1)$ because the definition occupies a constant drown area (Figure 2.3) or a constant number of symbols in the Verilog description for any n .

The size, the complexity and the depth of this version of decoder is out of discussion because the order of the size can not be reduced under the number of outputs ($m = 2^n$), for complexity $O(1)$ is the minimal order of magnitude, and for depth $O(\log n)$ is optimal taking into account we applied the “divide & impera” rule to build the structure of the decoder.

Non-recursive description

An iterative structural version of the previous recursive constructive definition is possible, because the outputs of the two $DCD_{n/2}$ from Figure 2.3 are also 2-input AND circuits, the same as the circuits on the output level. In this case we can apply the associative rule, implementing the last two levels by only one level of 4-input ANDs. And so on, until the output level of the 2^n n -input ANDs is driven by n EDCDs. Now we have the decoder represented in Figure 2.4). Apparently it is a constant depth circuit, but if we take into account that the number of inputs in the AND gates is not constant, then the depth is given by the depth of an n -input gate which is in $O(\log n)$. Indeed, an n -input AND has an efficient implementation as a binary tree of 2-input ANDs.

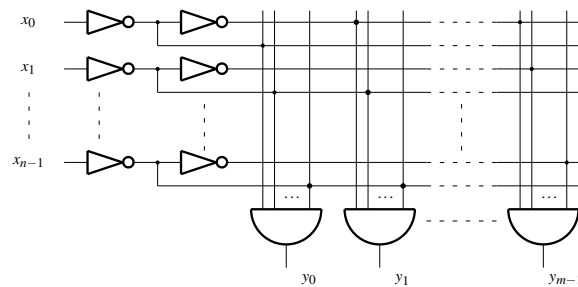


Figure 2.4: “Constant depth” DCD Applying the associative rule into the hierarchical network of AND_2 gates results the one level AND_n gates circuit driven by n EDCDs.

This “constant depth” DCD version – CDDCD – is faster than the previous for small values of n (usually for $n < 6$; for more details see Appendix **Basic circuits**), but the size becomes $S_{CDDCD}(n) = n \times 2^n + 2n \in O(n2^n)$. The price is over-dimensioned related to the gain, but for small circuits sometimes it can be accepted.

The pure structural description for DCD_3 is:

```

/* *****
File name:      dec3.v
Circuit name:   3-input Decoder
Description:    structural description of a 3-input decoder
***** */
module dec3(output [7:0] out,
            input [2:0] in );
    // internal connections
    wire in0, nin0, in1, nin1, in2, nin2;
    // EDCD for in[0]
    not not00(nin0, in[0]), not01(in0, nin0) ;
    // EDCD for in[1]
    not not10(nin1, in[1]), not11(in1, nin1) ;
    // EDCD for in[2]
    not not20(nin2, in[2]), not21(in2, nin2) ;
    // the second level
    and and0(out[0], nin2, nin1, nin0); // output 0
    and and1(out[1], nin2, nin1, in0 ); // output 1

```

```

and and2(out[2], nin2, in1, nin0); // output 2
and and3(out[3], nin2, in1, in0 ); // output 3
and and4(out[4], in2, nin1, nin0); // output 4
and and5(out[5], in2, nin1, in0 ); // output 5
and and6(out[6], in2, in1, nin0); // output 6
and and7(out[7], in2, in1, in0 ); // output 7
endmodule

```

For $n = 3$ the size of this iterative version is identical with the size which results from the recursive definition. There are meaningful differences only for big n . In real designs we do not need this kind of pure structural descriptions because the current synthesis tools manage very well even pure behavioral descriptions such that from the formal definition of the decoder.

Arithmetic interpretation

The decoder circuit is also an arithmetic circuit. It computes the numerical function of exponentiation: $Y = 2^X$. Indeed, for $n = i$ only the output y_i takes the value 1 and the rest of the outputs take the value 0. Then, the number represented by the binary configuration Y is 2^i .

Application

Because the expressions describing the m outputs of DCD_n are:

$$\begin{aligned}
 y_0 &= x'_{n-1} \cdot x'_{n-2} \cdot \dots \cdot x'_1 \cdot x'_0 \\
 y_1 &= x'_{n-1} \cdot x'_{n-2} \cdot \dots \cdot x'_1 \cdot x_0 \\
 y_2 &= x'_{n-1} \cdot x'_{n-2} \cdot \dots \cdot x_1 \cdot x'_0 \\
 &\dots \\
 y_{m-2} &= x_{n-1} \cdot x_{n-2} \cdot \dots \cdot x_1 \cdot x'_0 \\
 y_{m-1} &= x_{n-1} \cdot x_{n-2} \cdot \dots \cdot x_1 \cdot x_0
 \end{aligned}$$

the logic interpretation of these outputs is that they represent all the min-terms for an n -input function. Therefore, any n -input logic function can be implemented using a DCD_n and an *OR* with maximum $m - 1$ inputs.

Example 2.1 Let be the 3-input 2-output function defined in the table from Figure 2.5. A DCD_3 is used to compute all the min-terms of the 3 variables a , b , and c . A 3-input *OR* is used to “add” the min-terms for the function X , and a 4-input *OR* is used to “add” the min-terms for the function Y .

Each min-term is computed only once, but it can be used as many times as the implemented functions suppose.

◇

2.1.2 Demultiplexors

The structure of the decoder is included in the structure of the other usual circuits. Two of them are the *demultiplexor* circuit and the *multiplexer* circuit. These complementary functions are very important in digital systems because of their ability to perform “communication” functions. Indeed, demultiplexing

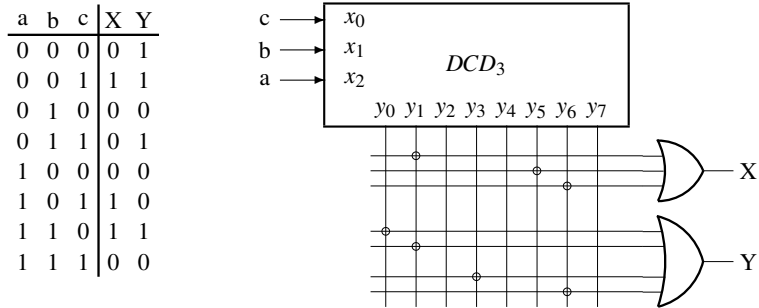


Figure 2.5:

means to spread a signal from a source to many destinations, selected by a binary code and multiplexing means the reverse operation to catch signals from distinct sources also selected using a selection code. Inside of both circuits there is a decoder used to identify the source of the signal or the destination of the signal by decoding the selection code.

Informal definition

The first informally described solution for implementing the function of an n -input demultiplexer is to use a decoder with the same number of inputs and m 2-input AND connected as in Figure 2.6. The value of the input *enable* is generated to the output of the gate opened by the activated output of the decoder DCD_n . It is obvious that a DCD_n is a $DMUX_n$ with $enable = 1$. Therefore, the size, depth of DMUXs are the same as for DCDs, because the depth is incremented by 1 and to the size is added a value which is in $O(2^n)$.

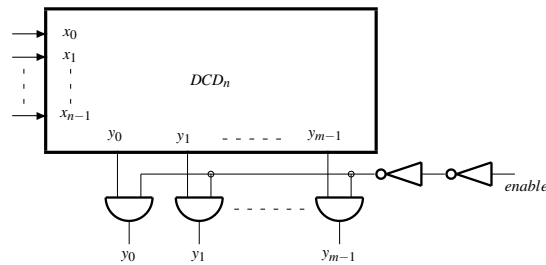


Figure 2.6: **Demultiplexor.** The n -input demultiplexer ($DMUX_n$) includes a DCD_n and 2^n AND_2 gates used to distribute the input *enable* in 2^n different places according to the n -bit selection code.

For example, if on the selection input $X = s$, then the outputs y_i take the value 0 for $i \neq s$ and $y_s = enable$. The inactive value on the outputs of this DMEX is 0.

Formal definition

Definition 2.3 The n -input demultiplexer – $DMUX_n$ – is a combinational circuit which transfers the 1-bit signal from the input *enable* to the one of the outputs y_{m-1}, \dots, y_0 selected by the n -bit selection code $X = x_{n-1}, \dots, x_0$, where $m = 2^n$. It has the following behavioral Verilog description:

```

/* *****
File name:      dmux.v
Circuit name:   Demultiplexor
Description:    behavioral description for a n-input demultiplexor
***** */
module dmux #(parameter inDim = n)(input  [inDim - 1:0]      sel    ,
                                     input  enable           ,
                                     output [(1 << inDim) - 1:0] out   );

    assign out = enable << sel;
endmodule

```

◇

Recursive definition

The DMUX circuit has also a recursive definition. The smallest DMUX, the elementary DMUX – EDMUX –, is a 2-output one, with a one-bit selection input. EDMUX is represented in Figure 2.7. It consists of an EDCD used to select, with its two outputs, the way for the signal *enable*. Thus, the EDMUX is a circuit that offers the possibility to transfer the same signal (*enable*) in two places (y_0 and y_1), according with the selection input (x_0) (see Figure 2.7).

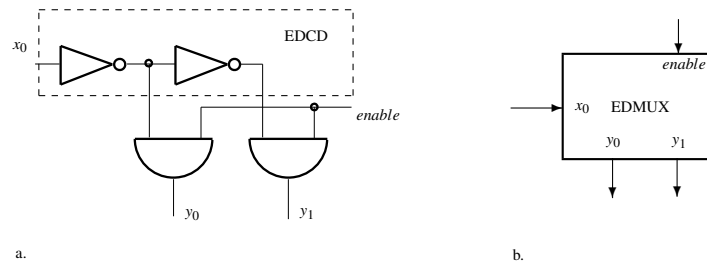


Figure 2.7: **The elementary demultiplexor.** **a.** The internal structure of an elementary demultiplexor (EDMUX) consists in an elementary decoder, 2 AND_2 gates, and an inverter circuit as input buffer. **b.** The logic symbol.

The same rule – *divide & impera* – is used to define an n -input demultiplexor, as follows:

Definition 2.4 $DMUX_n$ is defined as the structure represented in Figure 2.8, where the two $DMUX_{n-1}$ are used to select the outputs of an EDMUX. ◇

If the recursive rule is applied until the end the resulting circuit is a binary tree of EDMUXs. It has $S_{DMUX}(N) \in O(2^n)$ and $D_{DMUX}(n) \in O(n)$. If this depth is considered too big for the current application, the recursive process can be stopped at a convenient level and that level is implemented with a “constant depth” DMUXs made using “constant depth” DCDs. The mixed procedures are always the best. The previous definition is a suggestion for how to use small DMUXs to build bigger ones.

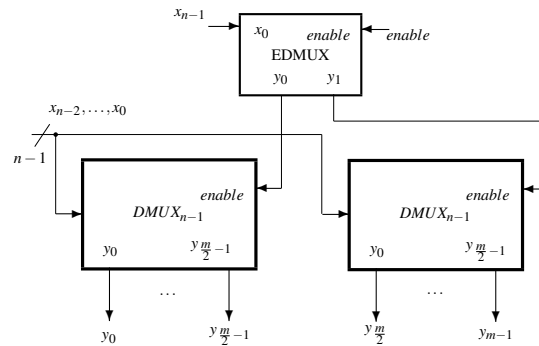


Figure 2.8: **The recursive definition of $DMUX_n$.** Applying the same rule for the two $DMUX_{n-1}$ a new level of 2 EDMUXs is added, and the output level is implemented using 4 $DMUX_{n-2}$. And so on until the output level is implemented using 2^{n-1} EDMUXs. The resulting circuit contains $2^n - 1$ EDMUXs.

2.1.3 Multiplexors

Now about the inverse function of demultiplexing: the **multiplexing**, i.e., to take a bit of information from a selected place and to send in one place. Instead of spreading by demultiplexing, now the multiplexing function gathers from many places in one place. Therefore, this function is also a communication function, allowing the interconnecting between distinct places in a digital system. In the same time, this circuit is very useful for implementing random, i.e. complex, logical functions, as we will see at the end of this chapter. More, in the next chapter we will see that the smallest multiplexor is used to build the basic memory circuits. Looks like this circuit is one of the most important basic circuit, and we must pay a lot of attention to it.

Informal definition

The direct intuitive implementation of a multiplexor with n selection bits – MUX_n – starts also from a DCD_n which is now serially connected with an AND-OR structure (see Figure 2.9). The outputs of the decoder open, for a given input code, only one AND gate that transfers to the output the corresponding selected input which, by turn, is OR-ed to the output y .

Applying in this structure the associativity rule, for the AND gates to the output of the decoder and the supplementary added ANDs, results the actual structure of MUX. The structure AND-OR maintains the size and the depth of MUX in the same orders as for DCD.

Formal definition

As for the previous two circuits – DCD and DMUX –, we can define the multiplexor using a behavioral (functional) description.

Definition 2.5 A multiplexer MUX_n is a combinational circuit having n selection inputs x_{n-1}, \dots, x_0 that selects to the output y one input from the $m = 2^n$ selectable inputs, i_{m-1}, \dots, i_0 . The Verilog description is:

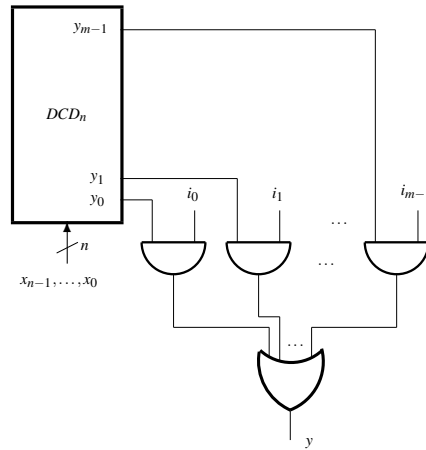


Figure 2.9: **Multiplexer.** The n selection inputs multiplexer MUX_n is made serial connecting a DCD_n with an AND-OR structure.

```

/* *****
File name:      mux.v
Circuit name:   Multiplexor
Description:    behavioral description for a n selection inputs
                multiplexor
***** */
module mux #(parameter inDim = n)
    (input [inDim-1:0]      sel , // selection inputs
     input [(1<<inDim)-1:0] in , // selected inputs
     output                out );
    assign out = in[ sel ];
endmodule

```

◇

The MUX is obviously a simple function. Its formal description, for any number of inputs has a constant size. The previous behavioral description is synthesizable efficiently by the current software tools.

Recursive definition

There is also a rule for composing large MUSs from the smaller ones. As usual, we start from an elementary structure. The elementary MUX – EMUX – is a *selector* that connects the signal i_1 or i_0 in y according to the value of the selection signal x_0 . The circuit is presented in Figure 2.10a, where an EDCD with the input x_0 opens only one of the two ANDs "added" by the OR circuit in y . Another version for EMUX uses *tristate* inverting drivers (see Figure 2.10c).

The definition of MUX_n starts from EMUX, in a recursive manner. This definition will show us that MUX is also a simple circuit ($C_{MUX}(n) \in O(1)$). In the same time this recursive definition will be a suggestion for the rule that composes big MUXs from the smaller ones.

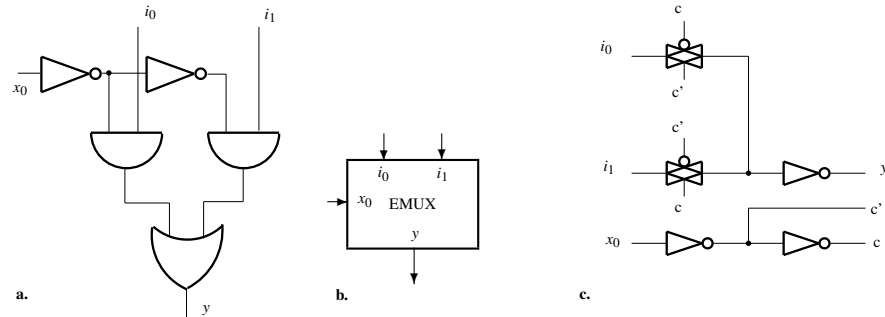


Figure 2.10: **The elementary multiplexer (EMUX).** **a.** The structure of EMUX containing an EDCCD and the smallest AND-OR structure. **b.** The logic symbol of EMUX. **c.** A version of EMUX using transmission gates (see section *Basic circuits*).

Definition 2.6 MUX_n can be made by serial connecting two parallel connected $MUX_{n/2}$ with an EMUX (see Figure 2.11 that is part of the definition), and $MUX_1 = EMUX$. \diamond

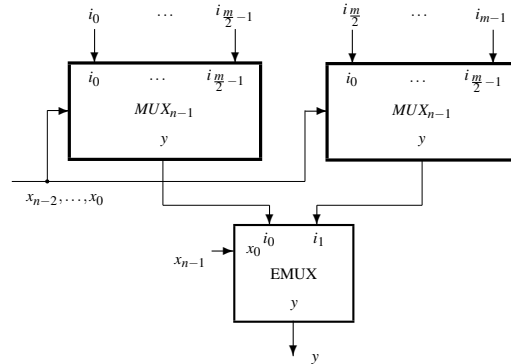


Figure 2.11: **The recursive definition of MUX_n .** Each MUX_{n-1} has a similar definition (two MUX_{n-2} and one EMUX), until the entire structure contains EMUXs. The resulting circuit is a binary tree of $2^n - 1$ EMUXs.

Structural aspects

This definition leads us to a circuit having the size in $O(2^n)$ (very good, because we have $m = 2^n$ inputs to be selected in y) and the depth in $O(n)$. In order to reduce the depth we can apply step by step the next procedure: for the first two levels in the tree of EMUXs we can write the equation

$$y = x_1(x_0i_3 + x'_0i_2) + x'_1(x_0i_1 + x'_0i_0)$$

that becomes

$$y = x_1x_0i_3 + x_1x'_0i_2 + x'_1x_0i_1 + x'_1x'_0i_0.$$

Using this procedure two or more levels (but not too many) of gates can be reduced to one. Carefully applied this procedure accelerate the speed of the circuit.

Application

Because the logic expression of a n selection inputs multiplexor is:

$$y = x_{n-1} \dots x_1 x_0 i_{m-1} + \dots + x'_{n-1} \dots x'_1 x_0 i_1 + x'_{n-1} \dots x'_1 x'_0 i_0$$

any n -input logic function is specified by the binary vector $\{i_{m-1}, \dots, i_1, i_0\}$. Thus any n input logic function can be implemented with a MUX_n having on its selected inputs the binary vector defining it.

Example 2.2 Let be function X defined in Figure 2.12 by its truth table. The implementation with a MUX_3 means to use the right side of the table as the defining binary vector.

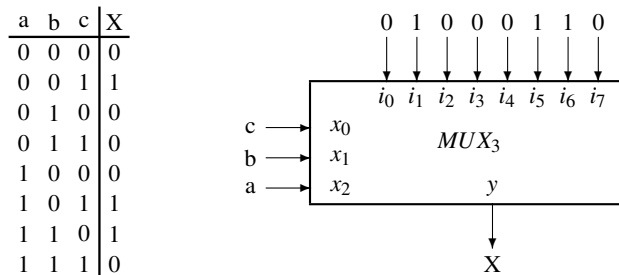


Figure 2.12:

◇

2.1.4 Priority encoder

An encoder is a circuit which connected to the outputs of a decoder provides the value applied on the input of the decoder. As we know only one output of a decoder is active at a time. Therefore, the encoder compute the index of the activated output. But, a real application of an encoder is to encode binary configurations provided by any kind of circuits. In this case, more than one input can be active and the encoder must have a well defined behavior. One of this behavior is to encode the most significant bit and to ignore the rest of bits. For this reason the encoder is a *priority encoder*.

The n -bit input, enabled priority encoder circuit, $PE(n)$, receives $x_{n-1}, x_{n-2}, \dots, x_0$ and, if the enable input is activated, $en = 1$, it generates the number $Y = y_{m-1}, y_{m-2}, \dots, y_0$, with $n = 2^m$, where Y is the biggest index associated with $x_i = 1$ if any, else zero output is activated. (For example: **if** $en = 1$, for $n = 8$, and $x_7, x_6, \dots, x_0 = 00110001$, **then** $y_2, y_1, y_0 = 101$ and $zero = 0$) The following Verilog code describe the behavior of $PE(n)$.

```

/*****
File name:      priority_encoder.v
Circuit name:   Priority Encoder
Description:    behavioral description of an 8-bit input priority encoder
*****/
module priority_encoder #(parameter m = 3)
    (input      [(1'b1<<m)-1:0] in      ,
     input      enable                ,
     output reg [m-1:0] out           ,
     output reg zero                  );
    integer i;
    always @(*) if (enable) begin out = 0;
                                for(i=(1'b1 << m)-1; i>=0; i=i-1)
                                    if ((out == 0) && in[i]) out = i;
                                if (in == 0) zero = 1;
                                    else zero = 0;
                                end
                                else begin out = 0;
                                        zero = 1;
                                end
endmodule

```

For testing the previous description the following test module is used:

```

/*****
File name:      .v
Circuit name:
Description:
*****/
module test_priority_encoder #(parameter m = 3);
  reg    [(1'b1<<m)-1:0]  in      ;
  reg    enable          ;
  wire  [m-1:0]         out     ;
  wire  zero            ;
  initial   begin       enable = 0;
                                     in  = 8'b11111111;
                                     #1 enable = 1;
                                     #1 in  = 8'b00000001;
                                     #1 in  = 8'b0000001x;
                                     #1 in  = 8'b000001xx;
                                     #1 in  = 8'b00001xxx;
                                     #1 in  = 8'b0001xxxx;
                                     #1 in  = 8'b001xxxxx;
                                     #1 in  = 8'b01xxxxxx;
                                     #1 in  = 8'b1xxxxxxx;
                                     #1 in  = 8'b110;
                                     #1 $stop;
                                     end
  priority_encoder dut(in      ,
                       enable  ,
                       out     ,
                       zero    );
  initial $monitor ($time , "enable=%b_in=%b_out=%b_zero=%b" ,
                    enable , in , out , zero);
endmodule

```

Running the previous code the simulation provides the following result:

```

time = 0    enable = 0  in = 11111111    out = 000    zero = 1
time = 1    enable = 1  in = 11111111    out = 111    zero = 0
time = 2    enable = 1  in = 00000001    out = 000    zero = 0
time = 3    enable = 1  in = 0000001x    out = 001    zero = 0
time = 4    enable = 1  in = 000001xx    out = 010    zero = 0
time = 5    enable = 1  in = 00001xxx    out = 011    zero = 0
time = 6    enable = 1  in = 0001xxxx    out = 100    zero = 0
time = 7    enable = 1  in = 001xxxxx    out = 101    zero = 0
time = 8    enable = 1  in = 01xxxxxx    out = 110    zero = 0
time = 9    enable = 1  in = 1xxxxxxx    out = 111    zero = 0
time =10    enable = 1  in = 00000110    out = 010    zero = 0

```

It is obvious that this circuit computes the integer part of the base 2 logarithm. The output *zero* is

used to notify that the input value is inappropriate for computing the logarithm, and “prevent” us from taking into account the output value.

2.1.5 Increment circuit

The simplest arithmetic operation is the increment. The combinational circuit performing this function receives an n -bit number, x_{n-1}, \dots, x_0 , and a one-bit command, inc , enabling the operation. The outputs, y_{n-1}, \dots, y_0 , and cr_{n-1} behaves according to the value of the command:

If $inc = 1$, then

$$\{cr_{n-1}, y_{n-1}, \dots, y_0\} = \{x_{n-1}, \dots, x_0\} + 1$$

else

$$\{cr_{n-1}, y_{n-1}, \dots, y_0\} = \{0, x_{n-1}, \dots, x_0\}.$$

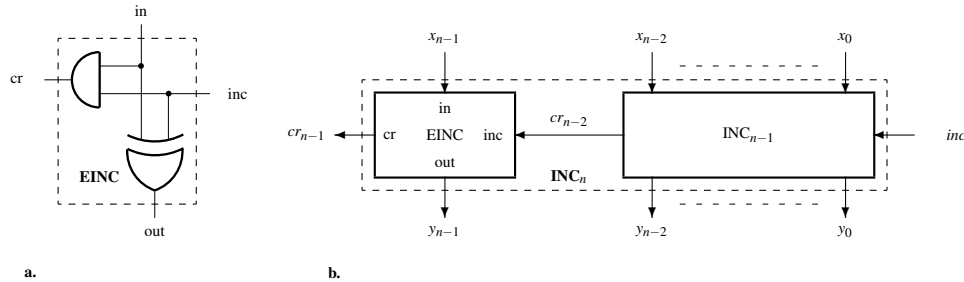


Figure 2.13: **Increment circuit.** **a.** The elementary increment circuit (called also **half adder**). **b.** The recursive definition for an n -bit increment circuit.

The increment circuit is built using as “brick” the **elementary increment circuit**, EINC, represented in Figure 2.13a, where the XOR circuit generate the increment of the input if $inc = 1$ (the current bit is complemented), and the circuit AND generate the carry for the the next binary order (if the current bit is incremented **and** it has the value 1). An n -bit increment circuit, INC_n is recursively defined in Figure 2.13b: INC_n is composed using an INC_{n-1} serially connected with an EINC, where $INC_0 = EINC$.

2.1.6 Adders

Another usual digital functions is the **sum**. The circuit associated to this function can be also made starting from a small elementary circuits, which adds two one-bit numbers, and looking for a simple recursive definitions for n -bit numbers.

The elementary structure is the well known *full adder* which consists in two half adders and an OR_2 . An n -bit adder could be done in a recursive manner as the following definition says.

Definition 2.7 *The full adder, FA, is a circuit which adds three 1-bit numbers generating a 2-bit result:*

$$FA(in1, in2, in3) = \{out1, out0\}$$

FA is used to build n -bit adders. For this purpose its connections are interpreted as follows:

- in_1, in_2 represent the i -th bits of two numbers
- in_3 represents the carry signal generated by the $i - 1$ stage of the addition process
- out_0 represents the i -th bit of the result
- out_1 represents the carry generated for the $i + 1$ -th stage of the addition process

Follows the Verilog description:

```

/*****
File name:      full_adder.v
Circuit name:   Full Adder
Description:    behavioral description of a full adder
*****/
module full_adder(output sum, carry_out, input in1, in2, carry_in);
    half_adder hal(sum1, carry1, in1, in2),
                ha2(sum, carry2, sum1, carry_in);
    assign carry_out = carry1 | carry2;
endmodule

```

```

/*****
File name:      half_adder.v
Circuit name:   Half Adder
Description:    behavioral description of a half adder
*****/
module half_adder(output sum, carry, input in1, in2);
    assign sum = in1 ^ in2,
            carry = in1 & in2;
endmodule

```

◇

Note: The half adder circuit is also an elementary increment circuit (see Figure 2.13a).

Definition 2.8 The n -bits ripple carry adder, (ADD_n), is made by serial connecting on the carry chain an ADD_{n-1} with a FA (see Figure 2.14). ADD_1 is a full adder.

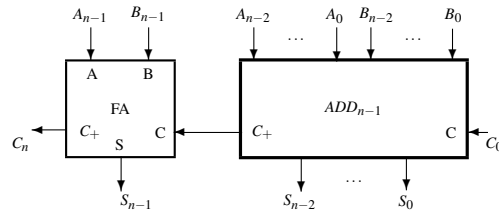


Figure 2.14: **The recursive defined n -bit ripple-carry adder (ADD_n).** ADD_n is simply designed adding to an ADD_{n-1} a full adder (FA), so as the carry signal ripples from one FA to the next.

```

/* *****
File name:      adder.v
Circuit name:   Adder
Description:    recursive structural description of a n-bit adder using
                the conditional generate statement
***** */
module adder #(parameter n = 4)(output [n-1:0] out ,
                                output cry ,
                                input [n-1:0] in1 ,
                                input [n-1:0] in2 ,
                                input cin );

    wire [n:1] carry ;
    assign cry = carry[n] ;
    generate
    if (n == 1) fullAdder firstAder (.out(out[0] ),
                                    .cry(carry[1] ),
                                    .in1(in1[0] ),
                                    .in2(in2[0] ),
                                    .cin(cin )
                                );
    else begin adder #(.n(n-1)) partAdder( .out(out[n-2:0] ),
                                            .cry(carry[n-1] ),
                                            .in1(in1[n-2:0] ),
                                            .in2(in2[n-2:0] ),
                                            .cin(cin )
                                        );
        fullAdder lastAdder (.out(out[n-1] ),
                            .cry(carry[n] ),
                            .in1(in1[n-1] ),
                            .in2(in2[n-1] ),
                            .cin(carry[n-1] )
                        );
    endgenerate
endmodule

```

```

/* *****
File name:      fullAdder.v
Circuit name:   Full Adder
Description:    behavioral description of a full adder
***** */
module fullAdder(  output  out, cry ,
                  input   in1, in2, cin);
    assign  cry = in1 & in2 | (in1 ^ in2) & cin ;
    assign  out = in1 ^ in2 ^ cin ;
endmodule

```

◇

The previous definition used the *conditioned generation* block.³ The Verilog code from the previous recursive definition can be used to simulate and to synthesize the adder circuit. For this simple circuit this definition is too sophisticated. It is presented here only to provide a simple example of how a recursive definition is generated.

A simpler way to define an adder is provided in the next example where a *generate* block is used.

Example 2.3 Generated *n*-bit adder:

```

/* *****
File name:      add.v
Circuit name:   Adder
Description:    structural description of a n-input adder using the
                generate statemnt
***** */
module add #(parameter n=8)(  input   [n-1:0] in1, in2 ,
                              input   cIn ,
                              output  [n-1:0] out ,
                              output  cOut );

    wire   [n:0]  cr ;
    assign cr[0] = cIn ;
    assign cOut = cr[n] ;
    genvar  i ;
    generate for (i=0; i<n; i=i+1) begin: S
        fa adder(  .in1   (in1[i] ),
                  .in2   (in2[i] ),
                  .cIn   (cr[i] ),
                  .out   (out[i] ),
                  .cOut  (cr[i+1])); end

    endgenerate
endmodule

```

³The use of the conditioned generation block for recursive definition was suggested to me by my colleague Radu Hobincu.


```

/* *****
File name:      fa.v
Circuit name:   Full Adder
Description:    structural description of a full adder
***** */
module fa( input  in1 , in2 , cIn  ,
           output out , cOut  );
  wire  xr  ;
  assign xr = in1 ^ in2 ;
  assign out = xr ^ cIn ;
  assign cOut = in1 & in2 | cIn & xr ;
endmodule

```

◇

Because the add function is very frequently used, the synthesis and simulation tools are able to “understand” the simplest one-line behavioral description used in the following module:

```

/* *****
File name:      add.v
Circuit name:   Adder
Description:    behavioral description of an adder
***** */
module add #(parameter n=8)( input  [n-1:0] in1 , in2 ,
                             input  cIn  ,
                             output [n-1:0] out  ,
                             output cOut );
  assign {cOut, out} = in1 + in2 + cIn ;
endmodule

```

Carry-Look-Ahead Adder

The size of ADD_n is in $O(n)$ and the depth is unfortunately in the same order of magnitude. For improving the speed of this very important circuit there was found a way for accelerating the computation of the carry: the *carry-look-ahead adder* (CLA_n). The fast carry-look-ahead adder can be made using a carry-look-ahead (CL) circuit for fast computing all the carry signals C_i and for each bit an half adder and a XOR (the modulo two adder)(see Figure 2.15). The half adder has two roles in the structure:

- sums the bits A_i and B_i on the output S
- computes the signals G_i (that *generates* carry as a local effect) and P_i (that allows the *propagation* of the carry signal through the binary level i) on the outputs CR and P .

The XOR gate adds modulo 2 the value of the carry signal C_i to the sum S .

In order to compute the carry input for each binary order an additional fast circuit must be build: the *carry-look-ahead circuit*. The equations describing it start from the next rule: *the carry toward the level*

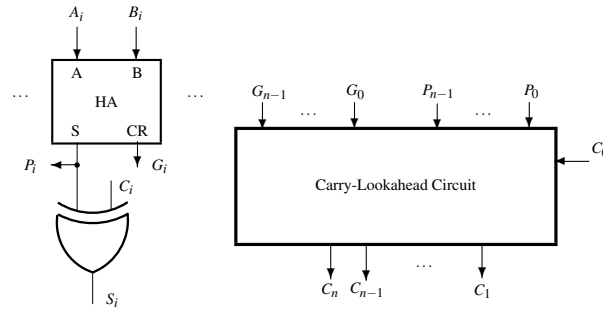


Figure 2.15: **The fast n -bit adder.** The n -bit Carry-Lookahead Adder (CLA_n) consists in n HAs, n 2-input XORs and the Carry-Lookahead Circuit used to compute faster the n C_i , for $i = 1, 2, \dots, n$.

$(i + 1)$ is **generated** if both A_i and B_i inputs are 1 or is **propagated** from the previous level if only one of A_i or B_i are 1. Results:

$$C_{i+1} = A_i B_i + (A_i + B_i) C_i = A_i B_i + (A_i \oplus B_i) C_i = G_i + P_i C_i.$$

Applying the previous rule we obtain the general form of C_{i+1} :

$$C_{i+1} = G_i + P_i G_{i-1} + P_i P_{i-1} G_{i-2} + P_i P_{i-1} P_{i-2} G_{i-3} + \dots + P_i P_{i-1} \dots P_1 P_0 C_0$$

for $i = 0, \dots, n$.

Computing the size of the carry-look-ahead circuit results $S_{CL}(n) \in O(n^3)$, and the theoretical depth is only 2. But, for real circuits an n -input gates can not be considered as a one-level circuit. In *Basic circuits* appendix (see section *Many-Input Gates*) is shown that an optimal implementation of an n -input simple gate is realized as a binary tree of 2-input gates having the depth in $O(\log n)$. Therefore, in a real implementation the depth of a carry-look ahead circuit has $D_{CLA} \in O(\log n)$.

For small n the solution with carry-look-ahead circuit works very good. But for larger n the two solutions, without carry-look-ahead circuit and with carry-look-ahead circuit, must be combined in many fashions in order to obtain a good price/performance ratio. For example, the ripple carry version of ADD_n is divided in two equal sections and two carry look-ahead circuits are built for each, resulting two serial connected $CLA_{n/2}$. The state of the art in this domain is presented in [Omondi '94].

It is obvious that the adder is a simple circuit. There exist constant sized definition for all the variants of adders.

2.1.7 Arithmetic and Logic Unit

All the before presented circuits have had associated only one logic or one arithmetic function. Now is the time to design the internal structure of a previously defined circuit having many functions, which can be selected using a selection code: the *arithmetic and logic unit* – ALU. ALU is the main circuit in any computational device, such as processors, controllers or embedded computation structures.

A generic version of a simple ALU is presented in the following example.

Example 2.4 *The 8-function ALU working on 32-bit numbers is described by the following Verilog module:*

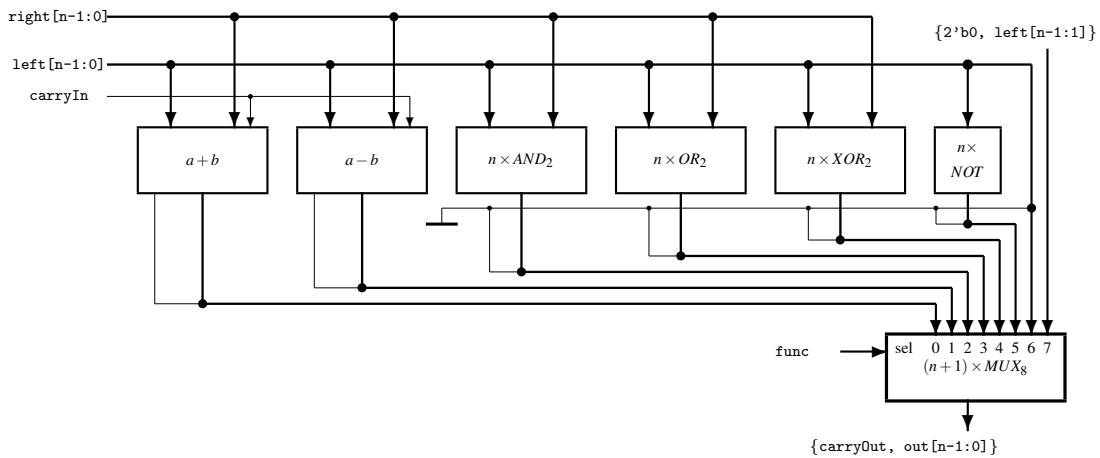


Figure 2.16: **The internal structure of the speculative version of an arithmetic and logic unit.** Each function is performed by a specific circuit and the output multiplexer selects the desired result.

```

/* *****
File name:      alu.v
Circuit name:   arithmetic and logic unit
Description:    the circuit selects, using the selection code 'func', one
                of the 8 functions
***** */
module ALU(input      carryIn      ,
           input      [2:0]   func  ,
           input      [31:0]  left , right ,
           output reg  carryOut  ,
           output reg  [31:0]  out   );
    always @(*)
    case (func)
        3'b000: {carryOut, out} = left + right + carryIn; //add
        3'b001: {carryOut, out} = left - right - carryIn; //sub
        3'b010: {carryOut, out} = {1'b0, left & right}; //and
        3'b011: {carryOut, out} = {1'b0, left | right}; //or
        3'b100: {carryOut, out} = {1'b0, left ^ right}; //xor
        3'b101: {carryOut, out} = {1'b0, ~left}; //not
        3'b110: {carryOut, out} = {1'b0, left}; //left
        3'b111: {carryOut, out} = {1'b0, left >> 1}; //shr
        default {carryOut, out} = 33'b0 - 1'b1;
    endcase
endmodule

```

◇

The ALU circuit can be implemented in many forms. One of them is the *speculative* version (see Figure 2.16) described by the *Verilog* module from Example 2.4, where the case structure describes, in

fact, an 8-input multiplexor for 33-bit words. We call this version speculative because *all* the possible functions are computed in order to be all available to be select when the function code arrives to the func input of ALU. This approach is efficient when the operands are available quickly and the function to be performed “arrives” lately (because it is usually decoded from the instruction fetched from a program memory). The circuit “speculates” computing all the defined functions offering 8 results from which the func code selects one. (This approach will be useful for the ALU designed for the stack processor described in Chapter 10.)

The speculative version provides a fast version in some specific designs. The price is the big size of the resulting circuit (mainly because the arithmetic section contains an adder and a subtractor, instead a smaller circuit performing add or subtract according to a bit used to complement the right operand and the carryIn signal).

An area optimized solution is provided in the next example.

Example 2.5 *Let be the 32-bit ALU with 8 functions described in Example 2.8. The implementation will be done using an adder-subtractor circuit and a 1-bit slice for the logic functions. Results the following Verilog description:*

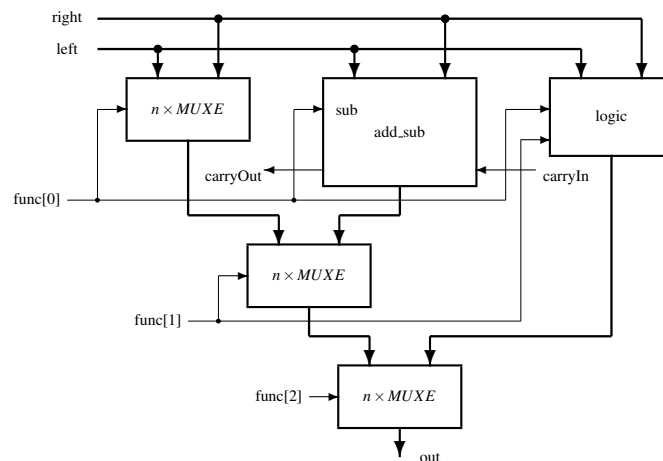


Figure 2.17: **The internal structure of an area optimized version of an ALU.** The `add_sub` module is smaller than an adder and a subtractor, but the operation “starts” only when `func[0]` is valid.

```

/* *****
File name:      structuralAlu.v
Circuit name:   ALU
Description:    structural description for
***** */
module structuralAlu(output  [31:0]  out      ,
                   output      carryOut ,
                   input      carryIn  ,
                   input  [31:0]  left   , right   ,
                   input  [2:0]   func   );
wire  [31:0]  shift , add_sub , arith , logic ;

addSub addSub(.out      (add_sub ),
              .cout      (carryOut),
              .left      (left   ),
              .right     (right   ),
              .cin       (carryIn ),
              .sub       (func[0] ));
logic log(.out      (logic   ),
          .left      (left   ),
          .right     (right   ),
          .op        (func[1:0]));
mux2     shiftMux(.out(shift      ),
                 .in0(left      ),
                 .in1({1'b0, left[31:1]}),
                 .sel(func[0]   )),
arithMux(.out(arith   ),
          .in0(shift   ),
          .in1(add_sub),
          .sel(func[1])),
outMux(.out(out      ),
        .in0(arith   ),
        .in1(logic   ),
        .sel(func[2]));

endmodule

```

```

/* *****
File name:      .v
Circuit name:
Description:
***** */
module mux2(input      sel      ,
            input  [31:0]  in0 , in1 ,
            output [31:0]  out   );
assign out = sel ? in1 : in0;
endmodule

```

```

/* *****
File name:      .v
Circuit name:
Description:
***** */
module addSub(output [31:0] out
              output   cout
              input  [31:0] left , right
              input   cin , sub
);
    assign {cout , out} = left + (right ^ {32{sub}}) + (cin ^ sub);
endmodule

```

```

/* *****
File name:      .v
Circuit name:
Description:
***** */
module logic(output reg [31:0] out
             input     [31:0] left , right
             input     [1:0]  op
);
    integer i;
    wire [3:0] f;
    assign f = {op[0], ~(~op[1] & op[0]), op[1], ~|op};
    always @(left or right or f)
        for (i=0; i<32; i=i+1) logicSlice(out[i], left[i], right[i], f);

    task    logicSlice;
        output    o;
        input     l , r;
        input    [3:0] f;
        o = f[{1,r}];
    endtask
endmodule

```

The resulting circuit is represented in Figure 2.17. This version can be synthesized on a smaller area, because the number of EMUXs is smaller, instead of an adder and a subtractor an adder/subtractor is used. The price for this improvement is a smaller speed. Indeed, the `add_sub` module “starts” to compute the addition or the subtract only when the signal `sub = func[0]` is received. Usually, the code `func` results from the decoding of the current operation to be performed, and, consequently, comes later. \diamond

We just learned a new feature of the Verilog language: how to use a `task` to describe a circuit used many times in implementing a simple, repetitive structure.

The internal structure of ALU consists mainly in n slices, one for each input pair `left[i]`, `right[i]` and a carry-look-ahead circuit(s) used for the arithmetic section. It is obvious that ALU is also a simple circuit. The magnitude order of the size of ALU is given by the size of the carry-look-ahead circuit because each slice has only a constant dimension and a constant depth. Therefore, the

fastest version implies a size in $O(n^3)$ because of the carry-look-ahead circuit. But, let's remind: the price for the fastest solution is always too big! For optimal solutions see [Omondi '94].

2.1.8 Comparator

Comparing functions are used in decisions. Numbers are compared to decide if they are equal or to indicate the biggest one. The n -bit comparator, $COMP_n$, is represented in Figure 2.18a. The numbers to be compared are the n -bit positive integers a and b . Three are the outputs of the circuit: lt_out , indicating by 1 that $a < b$, eq_out , indicating by 1 that $a = b$, and gt_out , indicating by 1 that $a > b$. Three additional inputs are used as expanding connections. On these inputs is provided information about the comparison done on the higher range, if needed. If no higher ranges of the number under comparison, then these three inputs must be connected as follows: $lt_in = 0$, $eq_in = 1$, $gt_in = 0$.

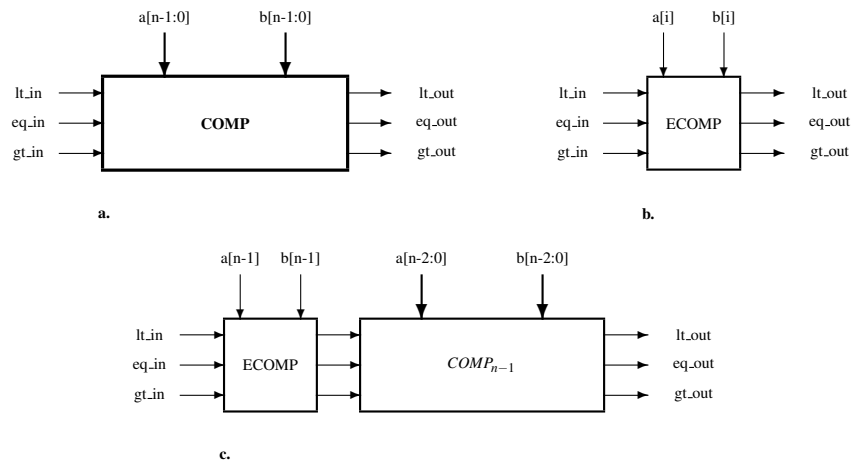


Figure 2.18: **The n -bit comparator, $COMP_n$.** **a.** The n -bit comparator. **b.** The elementary comparator. **c.** A recursive rule to built an $COMP_n$, serially connecting an $ECOMP$ with a $COMP_{n-1}$

The comparison is a numerical operation which starts inspecting the most significant bits of the numbers to be compared. If $a[n-1] = b[n-1]$, then the result of the comparison is given by comparing $a[n-2:0]$ with $b[n-2:0]$, else, the decision can be done comparing only $a[n-1]$ with $b[n-1]$ (using an elementary comparator, $ECOMP = COMP_1$ (see Figure 2.18b)), ignoring $a[n-2:0]$ and $b[n-2:0]$. Results a recursive definition for the comparator circuit.

Definition 2.9 An n -bit comparator, $COMP_n$, is obtained serially connecting an $COMP_1$ with a $COMP_{n-1}$. The Verilog code describing $COMP_1$ ($ECOMP$) follows:

```

/* *****
File name:      e_comp.v
Circuit name:   Elementary Comparator
Description:    behavioral description of an elementary comparator
***** */
module e_comp( input  a
               ,
               b
               ,
               lt_in  , // the previous e_comp decided lt
               eq_in  , // the previous e_comp decided eq
               gt_in  , // the previous e_comp decided gt
               output lt_out , // a < b
               eq_out , // a = b
               gt_out ); // a > b);
assign  lt_out = lt_in | eq_in & ~a & b,
        eq_out = eq_in & ~(a ^ b),
        gt_out = gt_in | eq_in & a & ~b;
endmodule

```

◇

The size and the depth of the circuit resulting from the previous definition are in $O(n)$. The size is very good, but the depth is too big for a high speed application.

An optimal comparator is defined using another recursive definition based on the *divide et impera* principle.

Definition 2.10 An n -bit comparator, $COMP_n$, is obtained using two $COMP_{n/2}$, to compare the higher and the lower half of the numbers (resulting $\{lt_out_high, eq_out_high, gt_out_high\}$ and $\{lt_out_low, eq_out_low, gt_out_low\}$), and a $COMP_1$ to compare gt_out_low with lt_out_low in the context of $\{lt_out_high, eq_out_high, gt_out_high\}$. The resulting circuit is represented in Figure 2.19. ◇

The resulting circuit is a \log -level binary tree of ECOMPs. The size remains in the same order⁴, but now the depth is in $O(\log n)$.

The bad news is: the HDL languages we have are unable to handle safely recursive definitions. The good news is: the synthesis tools provide good solutions for the comparison functions starting from a very simple behavioral description.

2.2 The many-output random circuit: Read Only Memory

The simple solution for the following many-output random circuits having the same inputs:

$$f(x_{n-1}, \dots, x_0)$$

⁴The actual size of the circuit can be minimized taking into account that: (1) the compared input of ECOMP cannot be both 1, (2) the output eq_out of one $COMP_{n/2}$ is unused, and (3) the expansion inputs of both $COMP_{n/2}$ are all connected to fix values.

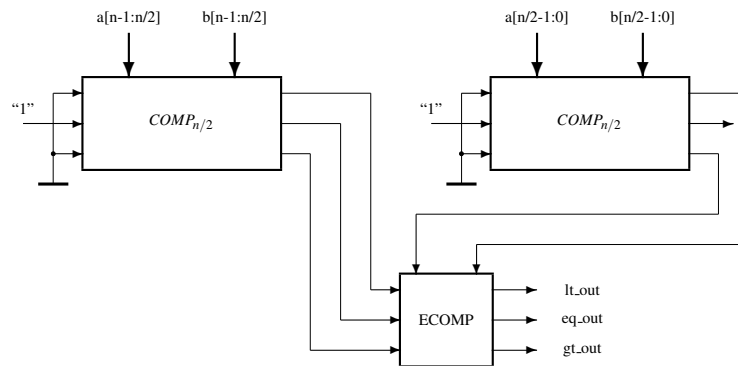


Figure 2.19: **The optimal n -bit comparator.** Applying the *divide et impera* principle a $COMP_n$ is built using two $COMP_{n/2}$ and an $ECOMP$. Results a \log -depth circuit with the size in $O(n)$.

$$\begin{aligned}
 &g(x_{n-1}, \dots, x_0) \\
 &\dots \\
 &s(x_{n-1}, \dots, x_0)
 \end{aligned}$$

is to connect in parallel many one-output circuits. The inefficiency of the solution become obvious when the structure of the MUX presented in Figure 2.9 is considered. Indeed, if we implement many MUXs with the same selection inputs, then the decoder DCD_n is replicated many time. One DCD is enough for many MUXs if the structure from Figure 2.20a is adopted. The DCD circuit is shared for implementing the functions f, g, \dots, s . The shared DCD is used to compute all possible *minterms* (see Appendix C.4) needed to compute an n -variable Boolean function.

Figure 2.20b is an example of using the generic structure from Figure 2.20a to implement a specific many-output function. Each output is defined by a different binary string. A 0 removes the associated AND, connecting the corresponding OR input to 0, and an 1 connects to the corresponding i -th input of each OR to the i -th DCD output. The equivalent resulting circuit is represented in Figure 2.20c, where some OR inputs are connected to *ground* and other directly to the DCD's output. Therefore, we use a technology allowing us to make "programmable" connections of some wires to other (each vertical line must be connected to one horizontal line). The *uniform* structure is "programmed" with a more or less *random* distribution of connections.

If De Morgan transformation is applied, the circuit from Figure 2.20c is transformed in the circuit represented in Figure 2.21a, where instead of an active high outputs DCD an active low outputs DCD is considered and the OR gates are substituted with NAND gates. The DCD's outputs are generated using NAND gates to *decode* the input binary word, the same as the gates used to *encode* the output binary word. Thus, a multi-output Boolean function works like a **trans-coder**. A trans-coder works translating all the binary input words into output binary words. The list of input words can be represented as an ordered list of sorted binary numbers starting with 0 and ending with $2^n - 1$. The table from Figure 2.22 represents the **truth table** for the multi-output function used to exemplify our approach. The left column contains all binary numbers from 0 (on the first line) until $2^n - 1 = 11 \dots 1$ (on the last line). In the right column the desired function is defined associating to each input an output. If the left column is an ordered list, the right column has a more or less random content (preferably more random for this type of solution).

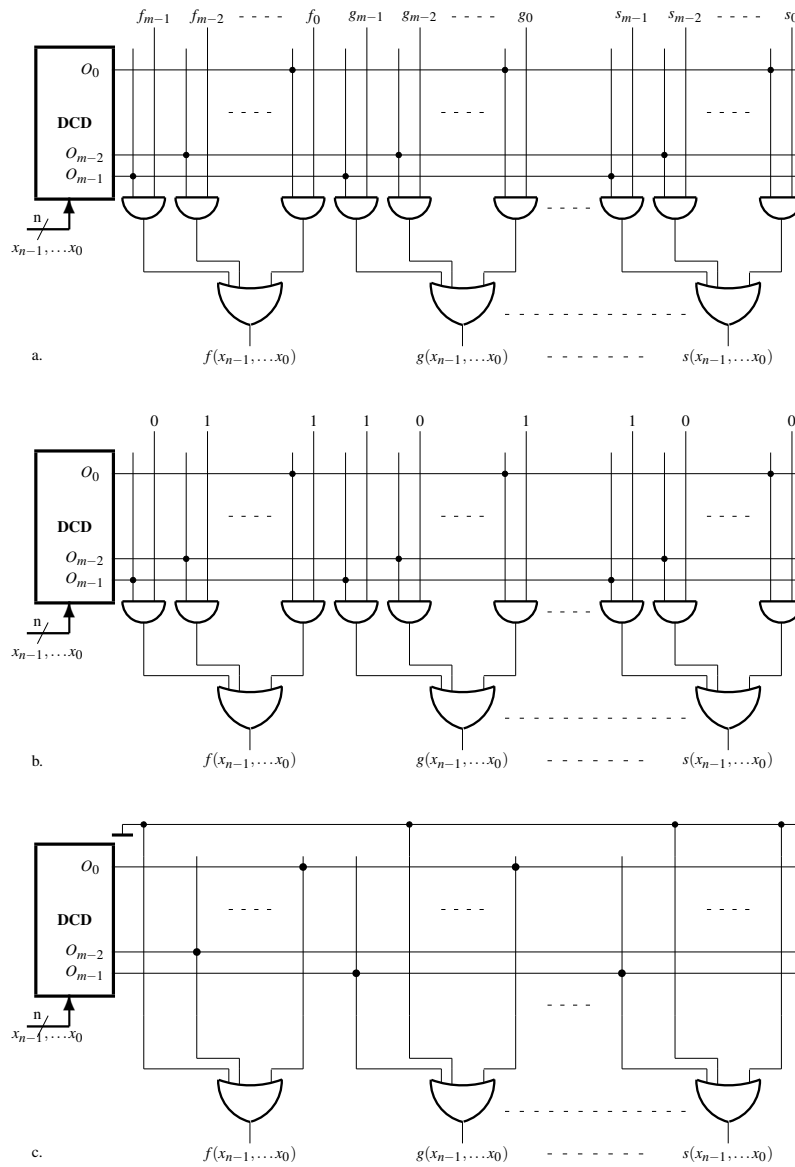


Figure 2.20: **Many-output random circuit.** **a.** One DCD and many AND-OR circuits. **b.** An example. **c.** The version using programmable connections.

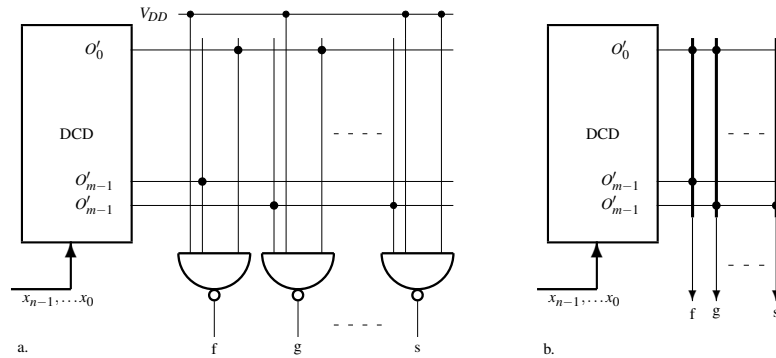


Figure 2.21: **The internal structure of a Read Only Memory used as trans-coder.** **a.** The internal structure. **b.** The simplified logic symbol where a thick vertical line is used to represent an m -input NAND gate.

Input	Output
00 ... 00	11 ... 0
...	...
11 ... 10	10 ... 0
11 ... 11	01 ... 1

Figure 2.22: **The truth table for a multi-output Boolean function.** The input rows can be seen as addresses, from 00...0 to 11...1 and the output columns as the content stored at the corresponding addresses.

The trans-coder circuit can be interpreted as a *fix content memory*. Indeed, it works like a memory containing at the location 00...00 the word 11...0, ... at the location 11...10 the word 10...0, and at the last location the word 01...1. The name of this kind of programmable device is **read only memory**, ROM.

Example 2.6 *The trans-coder from the binary coded decimal numbers to 7 segments display is a combinational circuit with 4 inputs, a, b, c, d , and 7 outputs A, B, C, D, E, F, G , each associated to one of the seven segments. Therefore we have to solve 7 functions of 4 variables (see the truth table from Figure 2.24).*

The Verilog code describing the circuit is:

```

/*****
File name:      even_segments.v
Circuit name:   Seven-Segment Transcoder
Description:    behavioral description of the seven-segment transcoder
*****/
module seven_segments( output reg [6:0] out ,
                    input [3:0] in );
    always @(in) case(in)
        4'b0000: out = 7'b0000001;
        4'b0001: out = 7'b1001111;
        4'b0010: out = 7'b0010010;
        4'b0011: out = 7'b0000110;
    endcase
endmodule

```

```

4'b0100: out = 7'b1001100;
4'b0101: out = 7'b0100100;
4'b0110: out = 7'b0100000;
4'b0111: out = 7'b0001111;
4'b1000: out = 7'b0000000;
4'b1001: out = 7'b0000100;
default out = 7'bxxxxxxx;
endcase
endmodule

```

The first solution is a 16-location of 7-bit words ROM (see Figure 2.23a. If inverted outputs are needed results the circuit from Figure 2.23b.

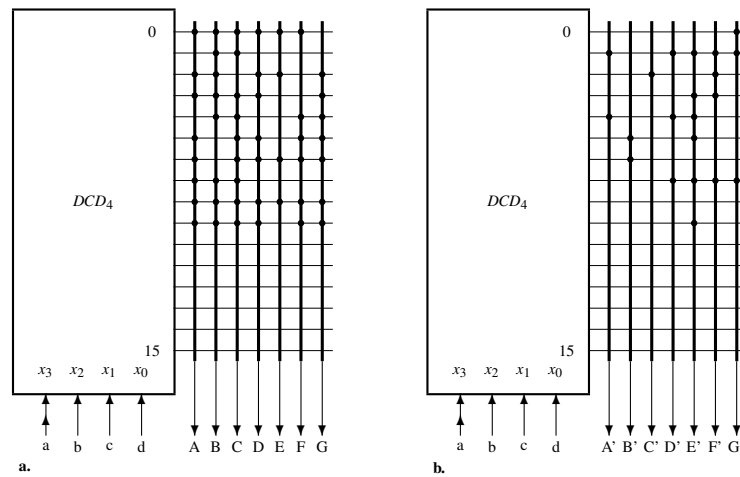


Figure 2.23: **The CLC as trans-coder designed serially connecting a DCD with an encoder.** Example: BCD to 7-segment trans-coder. **a.** The solution for non-inverting functions. **b.** The solution for inverting functions.

◇

2.3 Concluding about combinational circuits

The goal of this chapter was to introduce the main type of combinational circuits. Each presented circuit is important first, for its specific function and second, as a suggestion for how to build similar ones. There are a lot of important circuits undiscussed in this chapter. Some of them are introduced as problems at the end of this chapter.

Simple circuits vs. complex circuits Two very distinct class of combinational circuits are emphasized. The first contains simple circuits, the second contains complex circuits. The complexity of a circuit is distinct from the size of a circuit. Complexity of a circuit is given by the size of the definition used to specify that circuit. Simple circuits can achieve big sizes because they are defined using a repetitive pattern. A complex circuit can not be very big because its definition is dimensioned related with its size.

abcd	ABCDEFG
0000	1111110
0001	0110000
0010	1101101
0011	1111001
0100	0110011
0101	1011011
0110	1011111
0111	1110000
1000	1111111
1001	1111011
1010	-----
...
1111	-----

Figure 2.24: **The truth table for the 7 segment trans-coder.** Each binary represented decimal (in the left columns of inputs) has associated a 7-bit command (in the right columns of outputs) for the segments used for display. For unused input codes the output is “don’t care”.

Simple circuits have recursive definitions Each simple circuit is defined initially as an elementary module performing the needed function on the smallest input. Follows a recursive definition about how can be used the elementary circuit to define a circuit working for any input dimension. Therefore, any big simple circuit is a network of elementary modules which expands according to a specific rule. Unfortunately, the actual HDL, Verilog included, are not able to manage without (strong) restrictions recursive definitions neither in simulation nor in synthesis. The recursiveness is a property of simple circuits to be fully used only for our mental experiences.

Speeding circuits means increase their size Depth and size evolve in opposite directions. If the speed increases, the pay is done in size, which also increases. We agree to pay, but in digital systems the pay is not fair. We conjecture the bigger is performance the bigger is the unit price. Therefore, the pay increases more than the units we buy. It is like paying urgency tax. If the speed increases n times, then the size of the circuit increases more than n times, which is not fair but it is real life and we must obey.

Big sized complex circuits require programmable circuits There are software tolls for simulating and synthesizing complex circuits, but the control on what they generate is very low. A higher level of control we have using programmable circuits such as ROMs or PLA. PLA are efficient only if non-arithmetic functions are implemented. For arithmetic functions there are a lot of simple circuits to be used. ROM are efficient only if the randomness of the function is very high.

Circuits represent a strong but ineffective computational model Combinational circuits represent a *theoretical* solution for any Boolean function, but not an effective one. Circuits can do more than algorithms can describe. The price for their universal completeness is their ineffectiveness. In the general case, both the needed physical structure (a tree of EMUXs) and the symbolic specification (a binary string) increase exponentially with n (the number of binary input variables). More, in the general case only a *family of circuits* represents the solution.

To provide an effective computational tool new features must be added to a digital machine and some restrictions must be imposed on what is to be computable. The next chapters will propose improvements induced by successively closing appropriate loops inside the digital systems.

2.4 Problems

Gates

Problem 2.1 Determine the relation between the total number, N , of n -input m -output Boolean functions ($f : \{0, 1\}^n \rightarrow \{0, 1\}^m$) and the numbers n and m .

Problem 2.2 Let be a circuit implemented using 32 3-input AND gates. Using the appendix evaluate the area if 3-input gates are used and compare with a solution using 2-input gates. Analyze two cases: (1) the fan-out of each gate is 1, (2) the fan-out of each gate is 4.

Decoders

Problem 2.3 Draw DCD_4 according to Definition 2.9. Evaluate the area of the circuit, using the cell library from Appendix E, with the placement efficiency⁵ 70%. Estimate the maximum propagation time. The wires are considered enough short to be ignored their contribution in delaying signals.

Problem 2.4 Design a constant depth DCD_4 . Draw it. Evaluate the area and the maximum propagation time using the cell library from Appendix E. Compare the results with the results of the previous problem.

Problem 2.5 Propose a recursive definition for DCD_n using EDMUXs. Evaluate the size and the depth of the resulting structure.

Multiplexors

Problem 2.6 Draw MUX_4 using EMUX. Make the structural Verilog design for the resulting circuit. Organize the Verilog modules as hierarchical as possible. Design a tester and use it to test the circuit.

Problem 2.7 Define the 2-input XOR circuit using an EDCCD and an EMUX.

Problem 2.8 Make the Verilog behavioral description for a constant depth left shifter by maximum $m - 1$ positions for m -bit numbers, where $m = 2^n$. The “header” of the project is:

```

module left_shift( output [2m-2:0] out ,
                  input [m-1:0] in ,
                  input [n-1:0] shift );
    ...
endmodule

```

⁵For various reason the area used to place gates on Silicon can not completely used. Some unused spaces remain between gates. Area efficiency measures the degree of area use.

Problem 2.9 Make the Verilog structural description of a log-depth (the depth is $\log_2 16 = 4$) left shifter by 16 positions for 16-bit numbers. Draw the resulting circuit. Estimate the size and the depth comparing the results with a similar shifter designed using the solution of the previous problem.

Problem 2.10 Draw the circuit described by the Verilog module `leftRotate` in the subsection Shifters.

Problem 2.11 A barrel shifter for m -bit numbers is a circuit which rotate the bits the input word a number of positions indicated by the shift code. The “header” of the project is:

```

module barrel_shift(    output  [m-1:0] out    ,
                        input    [m-1:0] in    ,
                        input    [n-1:0] shift );
    ...
endmodule

```

Write a behavioral code and a minimal structural version in Verilog.

2.4.1 Recursive circuits

Problem 2.12 A comparator is circuit designed to compare two n -bit positive integers. Its definition is:

```

module comparator( input    [n-1:0] in1 , // first operand
                  input    [n-1:0] in2 , // second operand
                  output    eq    , // in1 = in2
                  output    lt    , // in1 < in2
                  output    gt    ); // in1 > in2
    ...
endmodule

```

1. write the behavioral description in Verilog
2. write a structural description optimized for size
3. design a tester which compare the results of the simulations of the two descriptions: the behavioral description and the structural description
4. design a version optimized for depth
5. define an expandable structure to be used in designing comparators for bigger numbers in two versions: (1) optimized for depth, (2) optimized for size.

Problem 2.13 Design a comparator for signed integers in two versions: (1) for negative numbers represented in $2s$ complement, (2) for negative numbers represented a sign and number.

Problem 2.14 Design an expandable priority encoder with minimal size starting from an elementary priority encoder, EPE, defined for $n = 2$. Evaluate its depth.

Problem 2.15 Design the Verilog structural descriptions for an 8-input adder in two versions: (1) using 8 FAs and a ripple carry connection, (2) using 8 HAs and a carry look ahead circuit. Evaluate both solutions using the cell library from Appendix E.

Problem 2.16 Design an expandable carry look-ahead adder starting from an elementary circuit.

Problem 2.17 Design an enabled incremter/decremter circuit for n -bit numbers. If $en = 1$, then the circuit increments the input value if $inc = 1$ or decrements the input value if $inc = 0$, else, if $en = 0$, the output value is equal with the input value.

Problem 2.18 Design an expandable adder/subtractor circuit for 16-bit numbers. The circuit has a carry input and a carry output to allow expandability. The 1-bit command input is sub . For $sub = 0$ the circuit performs addition, else it subtracts. Evaluate the area and the propagation time of the resulting circuit using the cell library from Appendix E.

2.4.2 Random circuits

Problem 2.19 The Gray counting means to count, starting from 0, so as at each step only one bit is changed. Example: the three-bit counting means 000, 001, 011, 010, 110, 111, 101, 100, 000, ... Design a circuit to convert the binary counting into the Gray counting for 8-bit numbers.

Problem 2.20 Design a converter from Gray counting to binary counting for n -bit numbers.

Problem 2.21 Write a Verilog structural description for ALU described in Example 2.3. Identify the longest path in the resulting circuit. Draw the circuit for $n = 8$.

Problem 2.22 Design in Verilog the behavioral and the structural description of a multiply and accumulate circuit, MACC, performing the function: $(a \times b) + c$, where a and b are 16-bit numbers and c is a 24-bit number.

Problem 2.23 Design the combinational circuit for computing

$$c = \sum_{i=0}^7 a_i \times b_i$$

where: a_i, b_i are 16-bit numbers. Optimize the size and the depth of the 8-number adder using a technique learned in one of the previous problem.

Problem 2.24 Exemplify the serial composition, the parallel composition and the serial-parallel composition in 0 order systems.

Problem 2.25 Write the logic equations for the BCD to 7-segment trans-coder circuit in both high active outputs version and low active outputs version. Minimize each of them individually. Minimize all of them globally.

Problem 2.26 Applying removing rules and reduction rules find the functions performed by 5-level universal circuit programmed by the following binary strings:

1. (0100)⁸

2. $(01000010)^4$
3. $(0100001011001010)^2$
4. $0^{24}(01000010)$
5. 00000001001001001111000011000011

Problem 2.27 Compute the biggest size and the biggest depth of an n -input, 1-output circuit implemented using the universal circuit.

2.5 Projects

Project 2.1 Finalize Project 1.1 using the knowledge acquired about the combinational structures in this chapter.

Project 2.2 Design a combinational floating point single precision (32 bit) multiplier according to the ANSI/IEEE Standard 754-1985, Standard for Binary Floating Point Arithmetic.

Chapter 3

MEMORIES:

First order, 1-loop digital systems

The magic images were placed on the wheel of the memory system to which correspondent other wheels on which were remembered all the physical contents of the terrestrial world – elements, stones, metals, herbs, and plants, animals, birds, and so on – and the whole sum of the human knowledge accumulated through the centuries through the images of one hundred and fifty great men and inventors. The possessor of this system thus rose above time and reflected the whole universe of nature and of man in his mind.

Frances A. Yates¹

A true memory is an associative one. Please do not confuse the physical support – the random access memory – with the function – the associative memory.

According to the mechanisms described in *Chapter 3* of this book, the step toward a new class of circuits means to close a new loop. This will be the first loop which closed over the combinational circuits already presented. Thus, a first degree of autonomy will be reached in digital systems: the *autonomy of the state of the circuit*. Indeed, the state of the circuit will be partially independent by the input signals, i.e., the output of the circuits do not depend on or not respond to certain input switching.

In this chapter we introduce some of the most important circuits used for building digital systems. The basic function in which they are involved is the *memory* function. Some events on the input of a memory circuit are significant for the state of the circuits and some are not. Thus, the circuit “memorizes”, by the state it reaches, the significant events and “ignores” the rest. The possibility to have an “attitude” against the input signals is given to the circuit by the autonomy induced by its internal loop. In fact, this first loop closed over a simple combinational circuit makes insignificant some input signals because the circuit is able to *compensate* their effect using the signals received back from its output.

The main circuits with one internal loop are:

¹She was Reader in the History of the Renaissance at the University of London. The quote is from *Giordano Bruno and the Hermetic Tradition*. Her other books include *The Art of Memory*.

- the **elementary latch** - the basic circuit in 1-OS, containing two appropriately loop-coupled gates; the circuit has two stable states being able to store 1 bit of information
- the **clocked latch** - the first digital circuit which accepts the clock signal as an input distinct from data inputs; the clock signal determines by its active **level** *when* the latch is triggered, while the data input determines *how* the latch switches
- the **master-slave** flip-flop - the *serial composition* in 1-OS, built by two clocked latches serially connected; results a circuit triggered by the active **transition** of clock
- the **random access memory (RAM)** - the *parallel composition* in 1-OS, containing a set of n clocked elementary latches accessed with a $DMUX_{\log_2 n}$ and a $MUX_{\log_2 n}$
- the **register** - the *serial-parallel composition* in 1-OS, made by parallel connecting master-slave flip-flops.

These first order circuits don't have a direct computational functionality, but are involved in supporting the following main processes in a computational machine:

- offer the storage support for implementing various memory functions (register files, stacks, queues, content addressable memories, associative memories, ...)
- are used for synchronizing different subsystems in a complex system (supports the pipeline mechanism, implements delay lines, stores the state of automata circuits).

3.1 Stable/Unstable Loops

There are two main types of loops closed over a combinational logic circuit: loops generating a stable behavior and loops generating an unstable behavior. We are interested in the first kind of loop that generates a *stable state* inside the circuit. The other loop cannot be used to build anything useful for computational purposes, except some low performance signal generators.

The distinction between the two types of loops is easily exemplified closing loops over the simplest circuit presented in the previous chapter, the elementary decoder (see Figure 3.1a).

The unstable loop is closed connecting the output y_0 of the elementary decoder to its input x_0 (see Figure 3.1b). Suppose that $y_0 = 0 = x_0$. After the time interval equal with t_{pLH} ² the output y_0 becomes 1. After another time interval equal with t_{pHL} the output y_0 becomes again 0. And so on, the two outputs of the decoder are *unstable* oscillating between 0 and 1 with a period of time $T_{osc} = t_{pLH} + t_{pHL}$, or the frequency $f_{osc} = 1/(t_{pLH} + t_{pHL})$.

The stable loop is obtained connecting the output y_1 of the elementary decoder to the input x_0 (see Figure 3.1c). If $y_1 = 0 = x_0$, then $y_0 = 1$ fixing again the value 0 to the output y_1 . If $y_1 = 1 = x_0$, then $y_0 = 0$ fixing again the value 1 to the output y_1 . Therefore, the circuit *has two stable states*. (For the moment we don't know how to switch from one state to another state, because the circuit has no input to command the switching from 0 to 1 or conversely. The solution comes soon.)

²the propagation time through the inverter when the output switches from the low logic level to the high level.

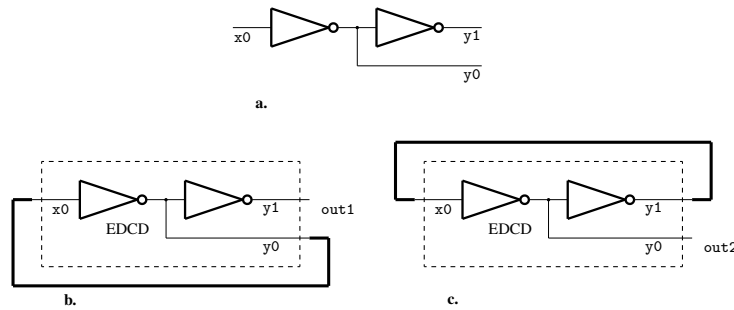


Figure 3.1: **The two loops closed over an elementary decoder.** **a.** The simplest combinational circuit: the one-input, elementary decoder. **b.** The unstable, inverting loop containing one (odd) inverting logic level(s). **c.** The stable, non-inverting loop containing two (even) inverting levels.

What is the main structural distinction between the two loops?

- The unstable loop has an *odd number of inverting levels*, thus the signal comes back to the output having the complementary value.
- The stable loop has an *even number of inverting levels*, thus the signal comes back to the output having the same value.

Example 3.1 Let be the circuit from Figure 3.2a, with 3 inverting levels on its internal loop. If the command input C is 0, then the loop is “opened”, i.e., the flow of the signal through the circular way is interrupted. If C switches in 1, then the behavior of the circuit is described by the wave forms represented in Figure 3.2b. The circuit generates a periodic signal with the period $T_{osc} = 3(t_{pLH} + t_{pHL})$ and frequency $f_{osc} = 1/3(t_{pLH} + t_{pHL})$. (To keep the example simple we consider that t_{pLH} and t_{pHL} have the same value for the three circuits.) \diamond

In order to be useful in digital applications, a loop closed over a combinational logic circuit must contain an even number of inverting levels *for all binary combinations applied to its inputs*. Else, for certain or for all input binary configurations, the circuit becomes unstable, useless for implementing computational functions. In the following, only even (in most of cases two) number of inverting levels are used for building the circuits belonging to 1-OS.

3.2 Elementary Structures

3.2.1 Elementary Latches

This chapter is devoted to introduce the elementary structure used to build memory systems: flip-flops, registers and random access memories. In order to be stable, all these elementary circuits have one loops with even (zero or two) inverting levels.

The reset-only latch is the *AND loop* circuit represented in Figure 3.3a. The *passive* input value for *AND loop* is 1 ($(\text{Reset})' = 1$), while the *active* input value is 0 ($(\text{Reset})' = 0$). If the passive input value is applied, then the output of the circuits is not affected (the output depends only by the other

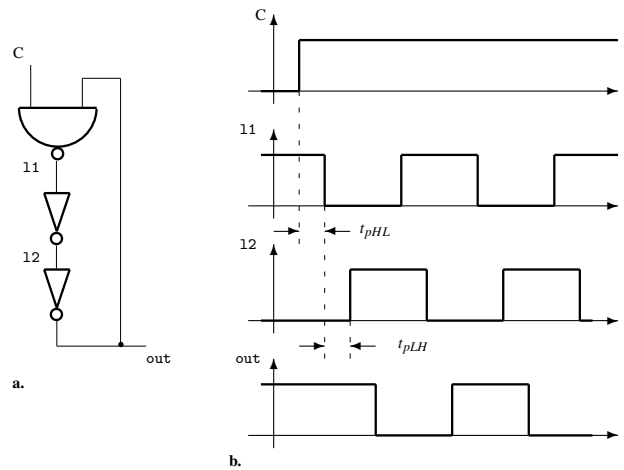


Figure 3.2: **The unstable loop.** The circuit version used for a low-cost and low-performance clock generator. **a.** The circuit with a three (odd) inverting circuits loop coupled. **b.** The wave forms drawn taking into account the propagation times associated to the low-high transitions (t_{pLH}) and to the high-low transitions (t_{pHL}).

input of the AND circuit). It can be 0 or 1, depending by the previous values applied on the input. When the active value is temporary applied, then the state of the circuit (the value of its output) switches in 0 and remains forever in this state, independent on the input value. We conclude that the circuit is sensitive to the signal 0 temporarily applied on its input, i.e., it is able to memorize forever the event 0.

The set-only latch is the *OR loop* circuit represented in Figure 3.3b. The *passive* value for *OR loop* is 0 ($\text{Set} = 0$) while the *active* input value is 1 ($\text{Set} = 1$). If the passive input value is applied, then the output of the circuits is not affected (the output depends only by the other input of the OR circuit). It can be 0 or 1, depending by the previous values applied on the input. When the active value is temporary applied, then the state of the circuit (the value of its output) switches in 1 and remains forever in this state, independent on the input value. We conclude that the circuit is sensitive to the signal 1 temporarily applied on its input, i.e., it is able to memorize forever the event 1.

The heterogenous set-reset latch results by combining the previous two latches (see Figure 3.3c). The circuit has zero inverting levels on the loop and two inputs: one active-low (active on 0) input, R' , to *reset* the circuit ($\text{out} = 0$) and another active-high (active on 1) input, S , to *set* the circuit ($\text{out} = 1$). The value 0 must remain to the input R' at least $2t_{pHL}$ for a stable switching of the circuit into the state 0, because the loop depth in the state 1 is given by the propagation time through both gates that switch from *high* to *low*. For a similar reason, the value 1 must remain to the input S at least $2t_{pLH}$ when the circuit must switch in 1.

The symmetric set-reset latch is obtained by applying De Morgan's law to the heterogenous elementary latch. In the first version, the OR circuit is transformed by De Morgan's law resulting the circuit from Figure 3.3d. The second version (see Figure 3.3e) is obtained applying the same law to the AND circuit. The passive input value for the NAND elementary latch is 1, while for the NOR elementary latch

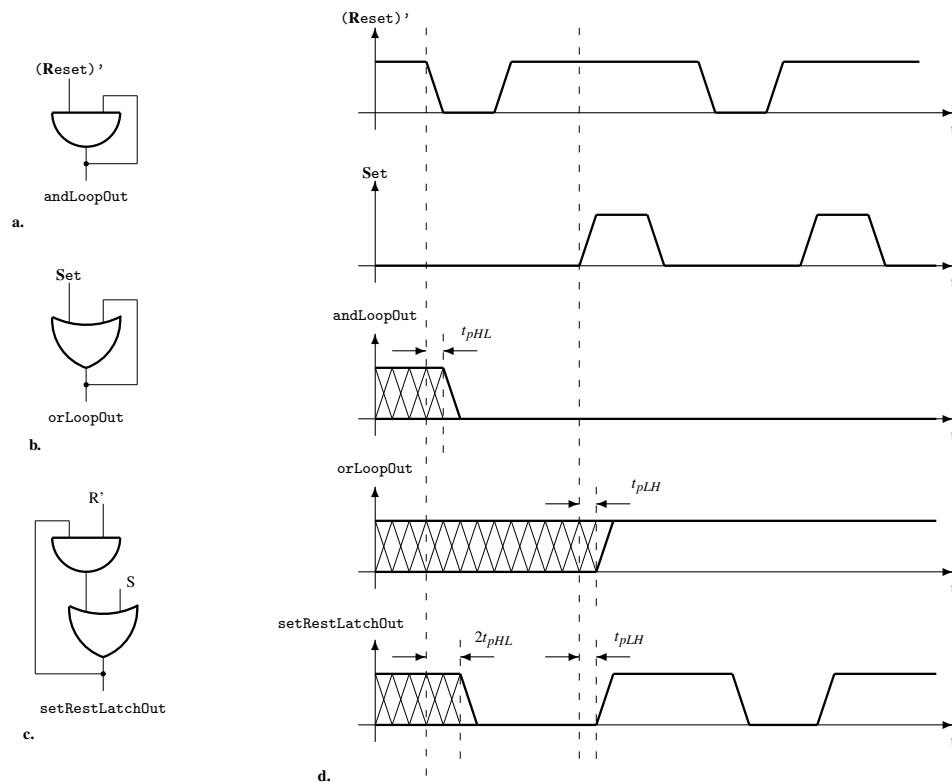


Figure 3.3: **The elementary latches.** Using the stable non-inverting loop (even inverting levels) elementary storage elements are built. **a.** *AND loop* provides a *reset-only* latch. **b.** *OR loop* provides the *set-only* version of a storage element. **c.** The heterogeneous elementary *set-reset* latch results combining the *reset-only* latch with the *set-only* latch. **d.** Symmetric elementary *NAND latch* with low-active commands S' and R' . **e.** Symmetric elementary *NOR latch* with high-active commands S and R .

it is 0. The active input value for the NAND elementary latch is 0, while for the NOR elementary latch it is 1. The symmetric structure of these latches have two outputs, Q and Q' .

VeriSim 3.1 *The Verilog description of NAND latch is:*

```
module elementary_latch(output out, not_out,
                       input not_set, not_reset);
    nand #2 nand0(out, not_out, not_set);
    nand #2 nand1(not_out, out, not_reset);
endmodule
```

For testing the behavior of the NAND latch just described, the following module is used:

```
module test_shortest_input;
    reg    not_set, not_reset;

    initial begin
        not_set    = 1;
        not_reset  = 1;
        #10 not_reset = 0;           // reset
        #10 not_reset = 1;
    end
endmodule
```

```

#10 not_set      = 0;          // set
#10 not_set      = 1;          // 1-st experiment
//#1 not_set     = 1;          // 2-nd experiment
//#2 not_set     = 1;          // 3-rd experiment
//#3 not_set     = 1;          // 4-th experiment
#10 not_set      = 0;          // another set
#10 not_set      = 1;
#10 not_reset   = 0;          // reset
#10 not_reset   = 1;
#10 $stop;

end

elementary_latch dut(out, not_out, not_set, not_reset);
endmodule

```

*In the first experiment the set signal is activated on 0 during 10ut (ut stands for unit time). In the second experiment (comment the line 9 and de-comment the line 10 of the test module), a set signal of 1ut is unable to switch the circuit. The third experiment, with 2ut set signal, generate an unstable simulated, but **non-actual**, behavior (to be explained by the reader). The fourth experiment, with 3ut set signal, determines the shortest set signal able to switch the latch (to be explained by the reader). \diamond*

In order to use these latches in more complex applications we must solve two problems.

The first latch problem : the inputs for indicating *how* the latch switches are the same as the inputs for indicating *when* the latch switches; we must find a solution for declutching the two actions building a version with distinct inputs for specifying “how” and “when”

The second latch problem : if we apply synchronously $S'=0$ and $R'=0$ on the inputs of NAND latch (or $S=1$ and $R=1$ on the inputs of OR latch), i.e., the latch is commanded “to switch in both states simultaneously”, then we can not predict what is the state of the latch after the ending of these two active signals.

The first latch problem will be partially solved in the next paragraph introducing the *clocked latch*, but the problem will be completely solved only by introducing the *master-slave* structure. The second latch problem will be solved only in the next chapter with the JK flip-flop, because the circuit needs more autonomy to “solve” the contradictory command that “says him” to switch in both states simultaneously. And, as we already know, more autonomy means at least a new loop.

Application: debouncing circuit Interfacing digital systems with the real world involves sometimes the use of mechanical switching contacts. The bad news is that this kind of contact does not provide an accurate transition. Usually when it closes, a lot of parasitic bounces come with the main transition (see wave forms S' and R' in Figure 3.4).

The debouncing circuit provide clean transitions when digital signals must generated by electro-mechanical switches. In Figure 3.4 an RS latch is used to clear up the bounces generated by a two-position electro-mechanical switch. The elementary latch latches the first transition from V_{DD} to 0. The bounces that follow have no effect on the output Q because the latch is already switched by the first transition in the state they intend to lead the circuit.

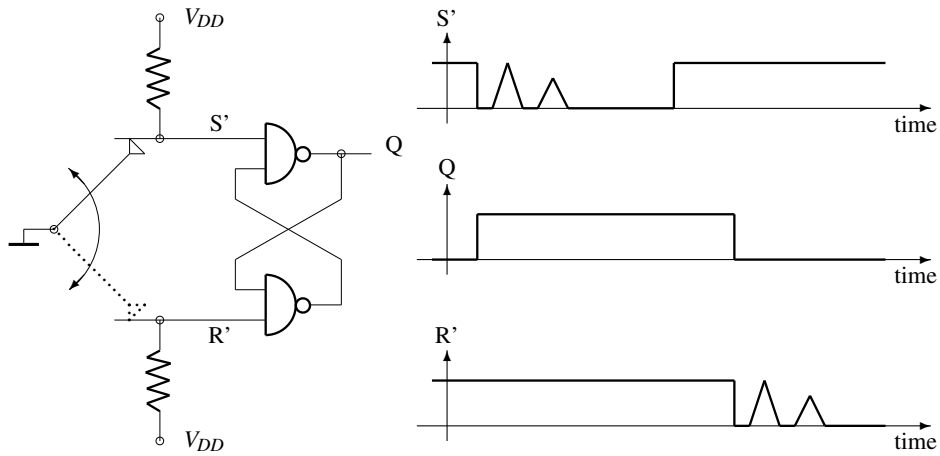


Figure 3.4: The debouncing circuit.

3.2.2 Elementary Clocked Latches

In order to start solving the *first latch problem* the elementary latch is supplemented with two gates used to validate the *data inputs* only during the active level of **clock**. Thus the *clocked elementary latch* is provided.

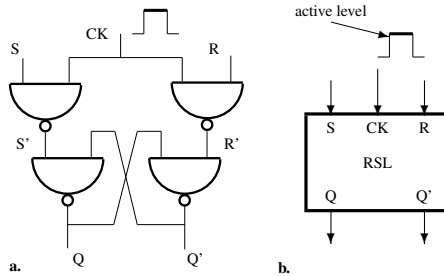


Figure 3.5: **Elementary clocked latch.** The transparent RS clocked latch is sensitive (transparent) to the input signals during the active level of the clock (the high level in this example). **a.** The internal structure. **b.** The logic symbol.

The NAND latch is used to exemplify (see Figure 3.5a) the *partial* separation between *how* and *when*. The signals R' and S' for the NAND latch are generated using two 2-input NAND gates. If the latch must be set, then on the input S we apply 1, R is maintained in 0 and, only *after that*, the clock is applied, i.e., the clock input CK switches temporary in 1. In this case the *active level* of the clock is the high level. For reset, the procedure is similar: the input R is activated, the input S is inactivated, and then the clock is applied.

We said that this approach allows only a *partial* declutching of *how* by *when* because on the active level of CK the latch is *transparent*, i.e., any change on the inputs S and R can modify the state of the circuit. Indeed, if $CK = 1$ and S or R is activated the latch is set or reset, and in this case *how* and *when*

are given only by the transition of these two signals, S for set or R for reset. The *transparency* will be avoided only when, in the next subsection, the transition of the output will be triggered by the active edge of clock.

The clocked latch does not solve the *second latch problem*, because for $R = S = 1$ the end of the active level of CK switches the latch in an unpredictable state.

VeriSim 3.2 *The following Verilog code can be used to understand how the elementary clocked latch works.*

```
module clocked_nand_latch(output out, not_out,
                        input set, reset, clock);
    elementary_latch the_latch(out, not_out, not_set, not_reset);
    nand #2 nand2(not_set, set, clock);
    nand #2 nand3(not_reset, reset, clock);
endmodule
```

◇

3.2.3 Data Latch

In the first order circuits class the second latch problem can be only avoided, **not removed**, defining a restriction on the input of the clocked latch. Indeed, introducing an inverter between the inputs of the RS clocked latch, as is shown in Figure 3.6a, the ambiguous command (simultaneous set and reset) can not be applied. We name the new input D (from **D**ata). Now, the situation $R = S = 1$ is avoided. The output is synchronized with the clock only if on the active level of CK the input D is stable.

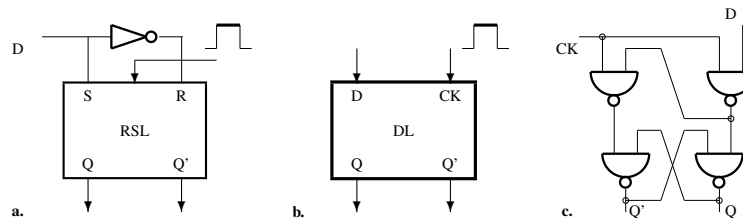


Figure 3.6: **The data latch.** Imposing the restriction $R = S'$ to an RS latch results the **D latch** without non-predictable transitions ($R = S = 1$ is not anymore possible). **a.** The structure. **b.** The logic symbol. **c.** An improved version for the data latch internal structure.

The output of this new circuit, called **D latch**, follows all the time the input D. Therefore, the autonomy of this circuit is questionable because act only in the time when the clock is inactive (on the inactive level of the clock). We say D latch is *transparent* on the active level of the clock signal, i.e, the output is sensitive to any input change during the active level of clock.

VeriSim 3.3 *The following Verilog code can be used to describe the behavior of a D latch.*

```
module data_latch( output reg out,
                  output not_out,
                  input data, clock);
    always @(data or clock) if (clock) out = data;
    assign not_out = ~out;
endmodule
```

◇

The main problem when data input D is separated by the timing input CK is the correlation between them. When this two inputs change in the same time, or, more precisely, during the same small time interval, some behavioral problems occur. In order to obtain a predictable behavior we must obey two important time restrictions: the *set-up time* and the *hold time*.

In Figure 3.6c an improved version of the circuit is presented. The number of components are minimized, the maximum depth of the circuit is maintained and the *fan-in* for the input D is reduced from 2 to 1.

VeriSim 3.4 *The following Verilog code can be used to understand how a D latch works.*

```

module test_data_latch;
  reg    data, clock;

  initial begin    clock = 0;
                  forever #10 clock = ~clock;
                end
  initial begin    data = 0;
                  #25 data = 1;
                  #10 data = 0;
                  #20 $stop;
                end
  data_latch dut(out, not_out, data, clock);
endmodule

module data_latch(output  out, not_out,
                  input   data, clock);

  not #2  data_inverter(not_data, data);

  clocked_nand_latch rs_latch(out, not_out, data, not_data, clock);
endmodule

```

The second initial construct from test_data_latch module can be used to apply data in different relation with the clock. ◇

The internal structure of the data latch (4 2-input NANDs and an inverter in Figure 3.6a) can be minimized opening the loop by disconnecting the output Q from the input of the gate generating Q' , and renaming it C . The resulting circuit is described by the following equation:

$$Q = ((D \cdot CK)' \cdot (C(D' \cdot CK)'))'$$

which can be successively transformed as follows:

$$Q = ((D \cdot CK) + (C(D' \cdot CK)'))$$

$$Q = ((D \cdot CK) + (C(D + CK')))$$

$$Q = D \cdot CK + C \cdot D + C \cdot CK' \text{ (anti - hasard redundancy)}$$

$$Q = D \cdot CK + C \cdot CK'$$

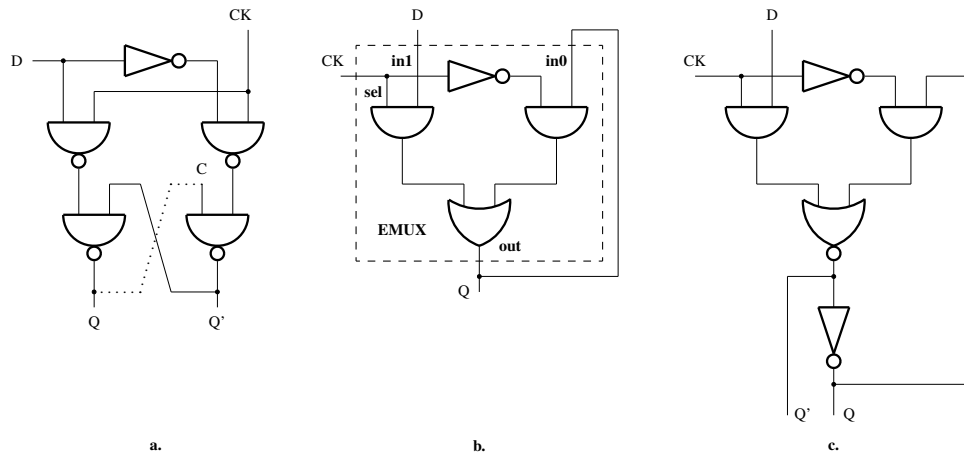


Figure 3.7: **The optimized data latch.** An optimized version is implemented closing the loop over an *elementary multiplexer*, EMUX. **a.** The resulting minimized structure for the circuit represented in Figure 3.6a. **b.** Implementing the minimized form using only inverting circuits.

The resulting circuit is an *elementary multiplexer* (the selection input is CK and the selected inputs are D , by $CK = 1$, and C , by $CK = 0$). Closing back the loop, by connecting Q to C , results the circuit represented in Figure 3.7a. The actual circuit has also the inverted output Q' and is implemented using only inverted gates as in Figure 3.7b. The circuit from Figure 3.6a (using the RSL circuit from Figure 3.5a) is implemented with 18 transistors, instead of 12 transistors supposed by the minimized form Figure 3.7b.

VeriSim 3.5 *The following Verilog code can be used as one of the shortest description for a D latch represented in Figure 3.7a.*

```

module mux_latch( output q
                 ,
                 input d, ck );
    assign q = ck ? d : q;
endmodule

```

In the previous module the assign statement, describing an elementary multiplexer, contains the loop. The variable q depends by itself. The code is synthesizable. \diamond

The *elementary decoder* was used to start the discussion about latches (see Figure 3.1). We ended using the *elementary multiplexer* to describe the most complex latch.

3.3 The Serial Composition: the Edge Triggered Flip-Flop

The first composition in 1-order systems is the *serial composition*, represented mainly by:

- the *master-slave* structure as the main mechanism that avoids the transparency of the storage structures

- the *delay flip-flop*, the basic storage circuit that allows to close the second loop in the synchronous digital systems
- the *serial register*, the first big and simple memory circuit having a recursive definition.

This class of circuits allows us to design synchronous digital systems. Starting from this point the inputs in a digital system are divided in two categories:

- clock inputs for synchronizing different parts of a digital system
- data and control inputs that receive the “informational” flow inside a digital system.

3.3.1 The Master-Slave Principle

In order to remove the transparency of the clocked latches, disconnecting completely the *how* from the *when*, the *master-slave principle* was introduced. This principle allows us to build a two state circuit named *flip-flop* that switches synchronized with the rising or falling *edge* of the clock signal.

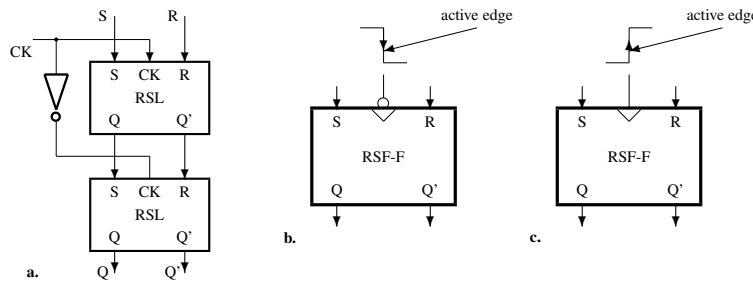


Figure 3.8: **The master-slave principle.** Serially connecting two RS latches, activated with different levels of the clock signal, results a non-transparent storage element. **a.** The structure of a RS master-slave flip-flop, active on the falling edge of the clock signal. **b.** The logic symbol of the RS flip-flop.

The principle consists in serially connecting two clocked latches and in applying the clock signal in opposite on the two latches (see Figure 3.8). In the exemplified embodiment the first latch is transparent on the high level of clock and the second latch is transparent on the low level of clock. (The symmetric situation is also possible: the first latch is transparent of the low level value of clock and the second no the high value of clock.) Therefore, there is no time interval in which the entire structure is transparent. In the first phase, $CK = 1$, the first latch is transparent - we call it the *master latch* - and it switches according to the inputs S and R. In the second phase $CK = 0$ the second latch - the *slave latch* - is transparent and it switches copying the state of the *master latch*. Thus the output of the entire structure is modified only synchronized with the negative transition of CK. We say the *RS master-slave flip-flop* switches with the falling (negative) edge of the clock. (The version triggered by the positive edge of clock is also possible.)

The switching moment of a master-slave structure is determined exclusively by the active edge of clock signal. Unlike the RS latch or data latch, which can sometimes be triggered (in the transparency time interval) by the transitions of the input data (R, S or D), the master-slave flip-flop flips only at the positive edge of clock (**always @ (posedge clock)**) or at the negative edge of clock (**always**

@(negedge clock)) edge of clock, according with the values applied on the inputs R and S. The *how* is now completely separated from the *when*. The *first latch problem* is finally solved.

VeriSim 3.6 *The following Verilog code can be used to understand how a master-slave flip-flop works.*

```

module master_slave(output out, not_out, input set, reset, clock);

    wire    master_out, not_master_out;

    clocked_nand_latch master_latch( .out    (master_out    ),
                                     .not_out(not_master_out ),
                                     .set    (set            ),
                                     .reset  (reset          ),
                                     .clock  (clock          )),
          slave_latch( .out    (out            ),
                      .not_out(not_out        ),
                      .set    (master_out     ),
                      .reset  (not_master_out ),
                      .clock  (~clock        ));

endmodule

◇

```

There are some other embodiments of the master-slave principle, but all suppose to connect latches serially.

Three very important time intervals must catch our attention in designing digital systems with edge triggered flip-flops:

set-up time – (t_{SU}) – the time interval before the active edge of clock in which the inputs R and S **must** stay unmodified allowing the correct switch of the flip-flop

edge transition time – (t_+ or t_-) – the positive or negative time transition of the clock signal (see Figure ??)

hold time – (t_H) – the time interval after the active edge of CK in which the inputs R and S **must** be stable (even if this time is zero or negative).

In the switching “moment”, that is approximated by the time interval $t_{SU} + t_+ + t_H$ or $t_{SU} + t_- + t_H$ “centered” on the active edge (+ or –), the data inputs must evidently be stable, because otherwise the flip-flop “does not know” what is the state in which “he” must switch.

Now, the problem of decoupling the *how* by the *when* is better solved. Although, this solution is not perfect, because the “moment” of the switch is approximated by the short time interval $t_{SU} + t_{+/-} + t_H$. But the “moment” does not exist for a digital designer. Always it must be a time interval, enough over-estimated for an accurate work of the designed machine.

3.3.2 The D Flip-Flop

Another tentative to remove the *second latch problem* leads to a solution that again avoids only the problem. Now the RS master-slave flip-flop is restricted to $R = S'$ (see Figure 3.9a). The new input is named also D, but now D means *delay*. Indeed, the flip-flop resulting by this restriction, besides avoiding the unforeseeable transition of the flip-flop, gains a very useful function: the output of the **D flip-flop**

follows the D input *with a delay of one clock cycle*. Figure 3.9c illustrates the delay effect of this kind of flip-flop.

Warning! *D latch* is a transparent circuit during the active level of the clock, unlike the *D flip-flop* which is no time transparent and switches only on the active edge of the clock.

VeriSim 3.7 *The structural Verilog description of a D flip-flop, provided only for simulation purpose, follows.*

```

module dff(output  out, not_out,
           input   d, clock );
  wire  not_d;

  not #2  data_inverter(not_d, d);

  master_slave  rs_ff(out, not_out, d, not_d, clock);
endmodule

```

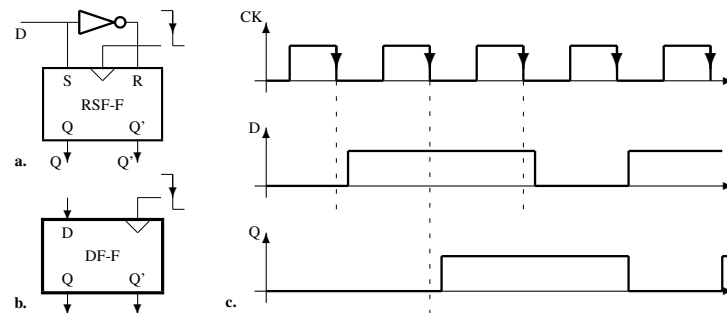


Figure 3.9: **The delay (D) flip-flop.** Restricting the two inputs of an RS flip-flop to $D = S = R'$, results an FF with predictable transitions. **a.** The structure. **b.** The logic symbol. **c.** The wave forms proving the delay effect of the D flip-flop.

The functional description currently used for a D flip-flop active on the negative edge of clock is:

```

module dff(output  reg out      ,
           input   d, clock);

  always @(negedge clock) out <= d;
endmodule

```

◇

The main difference between latches and flip-flops is that over the D flip-flop we can close a new loop in a very controllable fashion, unlike the D latch which allows a new loop, but the resulting behavior is not so controllable because of its transparency. Closing loops over D flip-flops result in synchronous systems. Closing loops over D latches result asynchronous systems. Both are useful, but in the first kind of systems the complexity is easiest manageable.

3.3.3 The Serial Register

Starting from the delay function of the last presented circuit (see Figure 3.9) a very important function and the associated structure can be defined: the *serial register*. It is very easy to give a recursive definition to this simple circuit.

Definition 3.1 An n -bit serial register, SR_n , is made by serially connecting a D flip-flop with an SR_{n-1} . SR_1 is a D flip-flop. \diamond

In Figure 3.10 is shown a SR_n . It is obvious that SR_n introduces a n clock cycle delay between its input and its output. The current application is for building digital controlled “delay lines”.

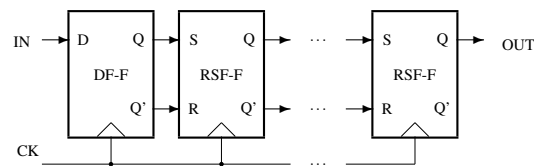


Figure 3.10: **The n -bit serial register (SR_n)**. Triggered by the active edge of the clock, the content of each RSF-F is loaded with the content of the previous RSF-F.

We hope that now it is very clear what is the role of the master-slave structure. Let us imagine a “serial register built with D latches”! The transparency of each element generates the strange situation in which at each clock cycle the input is loaded in a number of latches that depends by the length of the active level of the clock signal and by the propagation time through each latch. Results an uncontrolled system, useless for any application. Therefore, for controlling the propagation with the clock signal we *must* use the master-slave, non-transparent structure of D flip-flop that switches on the positive or negative edge of clock.

VeriSim 3.8 The functional description currently used for an n -bit serial register active on the positive edge of clock is:

```

/* *****
File name:      serial_register.v
Circuit name:   Serial register
Description:    behavioral description of a n-bit serial register
***** */
module serial_register #(parameter n = 1024)
    (output out
     ,
     input in, enable, clock);
    reg[0:n-1] serial_reg;

    assign out = serial_reg[n-1];
    always @(posedge clock)
        if (enable) serial_reg <= {in, serial_reg[0:n-2]};
endmodule

```

\diamond

3.4 The Parallel Composition: the Random Access Memory

The *parallel composition* in 1-OS provides the random access memory (RAM), which is the main storage support in digital systems. Both, data and programs are stored on this physical support in different forms. Usually we call these circuits improperly *memories*, even if the memory function is something more complex, which suppose besides a storage device a specific access mechanism for the stored information. A true memory is, for example, an *associative memory* (see the next subchapters about applications), or a stack memory (see next chapter).

This subchapter introduces two structures:

- a trivial composition, but a very useful circuit: the *n-bit latch*
- the asynchronous *random access memory* (RAM),

both involved in building big but simple recursive structures.

3.4.1 The *n*-Bit Latch

The *n*-bit latch, L_n , is made by parallel connecting n data latches clocked by the same CK. The system has n inputs and n outputs and stores an *n*-bit word. L_n is a *transparent* structure on the active level of the CK signal. The *n*-bit latch must be distinguished by the *n-bit register* (see the next section) that switches on the edge of the clock. In a synchronous digital system is forbidden to close a combinational loop over L_n .

VeriSim 3.9 A 16-bit latch is described in Verilog as follows:

```

/* *****
File name:      n_latch.v
Circuit name:   n-Bit Latch
Description:    behavioral description of a n-bit latch
***** */
module  n_latch #(parameter n = 16)(output  reg [n-1:0] out      ,
                                     input    [n-1:0] in        ,
                                     input    clock              );

    always @(in or clock)
        if (clock == 1) // the active-high clock version
        //if (clock == 0) // the active-low clock version
            out = in;
endmodule

```

◇

The n -bit latch works like a memory, storing n bits. The only deficiency of this circuit is due to the access mechanism. We must control the value applied on all n inputs when the latch changes its content. More, we can not use selectively the content of the latch. The two problems are solved adding some combinational circuits to limit both the changes and the use of the stored bits.

3.4.2 Asynchronous Random Access Memory

Adding combinational circuits for accessing in a more flexible way an m -bit latch for write and read operations, results one of the most important circuits in digital systems: the **random access memory**. This circuit is the biggest and simplest digital circuit. And we can say it can be the biggest *because* it is the simplest.

Definition 3.2 The m -bit random access memory, RAM_m , is a linear collection of m D (data) latches parallel connected, with the 1-bit common data inputs, DIN . Each latch receives the clock signal distributed by a $DMUX_{\log_2 m}$. Each latch is accessed for reading through a $MUX_{\log_2 m}$. The selection code is common for $DMUX$ and MUX and is represented by the p -bit address code: A_{p-1}, \dots, A_0 , where $p = \log_2 m$.

◇

The logic diagram associated with the previous definition is shown in Figure 3.11. Because no one of the input signal is clock related, this version of RAM is considered an asynchronous one. The signal WE' is the low-active *write enable* signal. For $WE' = 0$ the write operation is performed in the memory cell selected by the address A_{n-1}, \dots, A_0 .³ The wave forme describing the relation between the input and output signals of a RAM are represented in Figure 3.12, where the main time restrictions are the followings:

- t_{ACC} : access time - the propagation time from address input to data output when the read operation is performed; it is defined as a minimal value

³The actual implementation of this system uses optimized circuits for each 1-bit storage element and for the access circuits. See Appendix C for more details.)

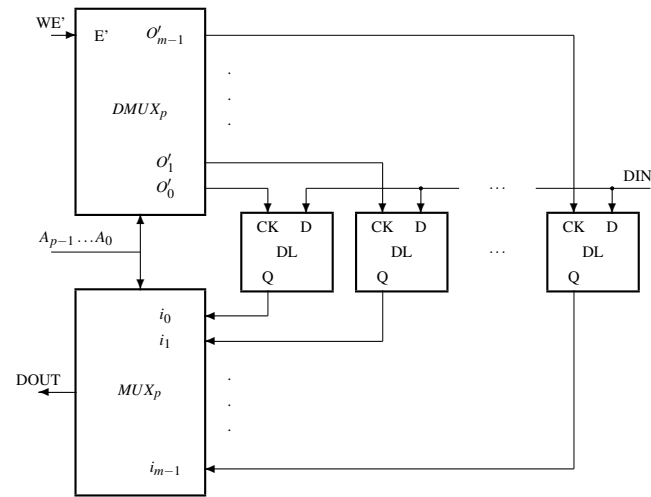


Figure 3.11: **The principle of the random access memory (RAM).** The clock is distributed by a DMUX to one of $m = 2^p$ DLs, and the data is selected by a MUX from one of the m DLs. Both, DMUX and MUX use as selection code a p -bit address. The one-bit data DIN can be stored in the clocked DL.

- t_W : write signal width - the length of active level of the write enable signal; it is defined as the shortest time interval for a secure writing
- t_{ASU} : address set-up time related to the occurrence of the write enable signal; it is defined as a minimal value for avoiding to disturb the content of other than the storing cell selected by the current address applied on the address inputs
- t_{AH} : address hold time related to the end transition of the write enable signal; it is defined as a minimal value for similar reasons
- t_{DSU} : data set-up time related to the end transition of the write enable signal; it is defined as a minimal value that ensure a proper writing
- t_{DH} : data hold time related to the end transition of the write enable signal; it is defined as a minimal value for similar reasons.

The just described version of a RAM represents only the *asynchronous core* of a memory subsystem, which must have a synchronous behavior in order to be easy integrated in a robust design. In Figure 3.11 there is no clock signal applied to the inputs of the RAM. In order to synchronize the behavior of this circuit with the external world, additional circuits must be added (see the first application in the next subchapter: *Synchronous RAM*).

The actual organization of an asynchronous RAM is more elaborated in order to provide the storage support for a big number of m -bit words.

VeriSim 3.10 *The functional description of a asynchronous $n = 2^p$ m -bit words RAM follows:*

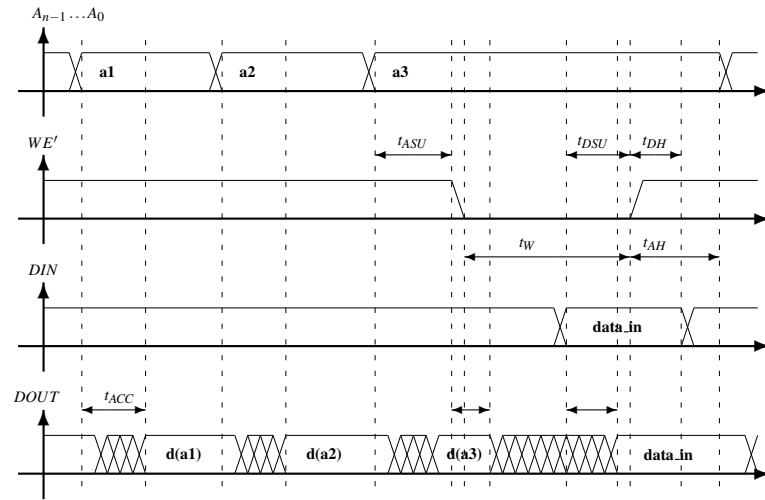


Figure 3.12: **Read and write cycles for an asynchronous RAM.** Reading is a combinational process of selecting. The access time, t_{ACC} , is given by the propagation through a big MUX. The write enable signal must be strictly included in the time interval when the address is stable (see t_{ASU} and t_{AH}). Data must be stable related to the positive transition of WE' (see t_{DSU} and t_{DH}).

```

/* *****
File name:      ram.v
Circuit name:   Asynchronous RAM
Description:    behavioral description of an asynchronous random-access
                memory
***** */
module ram(input  [m-1:0]  din ,           // data input
           input  [p-1:0]  addr ,        // address
           input                    we ,  // write enable
           output [m-1:0]  dout );       // data out
  reg [m-1:0] mem[(1'b1<<p)-1:0];        // the memory

  assign dout = mem[addr];               // reading

  always @(din or addr or we) if (we) mem[addr] = din; // writing
endmodule

```

◇

The real structural version of the storage array will be presented in two stages. First the number of bits per word will be expanded, then the solution for a big number of words number of words will be presented.

Expanding the number of bits per word

The pure logic description offered in Figure 3.11 must be reconsidered in order (1) to optimize it and (2) to show how the principle it describe can be used for designing a many-bit word RAM. The circuit structure from Figure 3.13 represents the m -bit word RAM. The circuit is organized in m columns, one for each bit of the m -bit word. The DMUX structure is shared all by the m columns, while each column has it own MUX structure. Let us remember that both, the DMUX and MUX circuits are structured around a DCD. See Figure 2.6 and 2.9, where the first level in both circuits is a decoder, followed by a linear network of 2-input ANDs for DMUX, and by an AND-OR circuit for MUX. Then, only one decoder, DCD_p , must be provided for the entire memory. It is shared by the demultiplexing function and by the m multiplexors. Indeed, the outputs of the decoder, $LINE_{n-1}, \dots, LINE_1, LINE_0$, are used to drive:

- one AND_2 gate associate cu each line in the array, whose output clocks the DL latches associated to one word; with these gates the decoder forms the demultiplexing circuit used to clock, when $WE = 1$, the latches selected (addressed) by the current value of the address: A_{p-1}, \dots, A_0
- m AND_2 gates, one in each column, selecting the read word to be ORed to the outputs $DOUT_{m-1}, DOUT_{m-2}, \dots, DOUT_0$; with the AND-OR circuit from each COLUMN the decoder forms the multiplexor circuit associated to each output bit of the memory.

The array of lathes is organized in n and m columns. Each line is driven for write by the output of a demultiplexer, while for the read function the addressed line (word) is selected by the output of a decoder. The output value is gathered from the array using m multiplexors.

The reading process is a pure combinational one, while the writing mechanism is an asynchronous sequential one. The relation between the WE signal and the address bits is very sensitive. Due to the combinational hazard to the output of DCD, the WE' signal must be activated only when the DCD's outputs are stabilized to the final value, i.e., t_{ASU} before the fall edge of WE' or t_H after the rise edge of WE'.

Expanding the number of words by two dimension addressing

The factor form on silicon of the memory described in Figure 3.13 is very unbalanced for $n \gg \gg m$. Expanding the number of words for the a RAM in the previous, one block version is not efficient because request a complex lay-out involving very long wires. We are looking for a more "squarish" version of the lay-out for a big memory. The solution is to connect in parallel many m -column blocks, thus defining a many-word from which to select one word using another level of multiplexing. The reading process selects the many-word containing the requested word from which the requested word is selected.

The internal organization of memory is now a two dimension array of rows and columns. Each row contains a many-word of 2^q words. Each column contains a number of 2^r words. The memory is addressed using the $(p = r + q)$ -bit address:

$$\text{addr}[p-1:0] = \{\text{rowAddr}[r-1:0], \text{colAddr}[q-1:0]\}$$

The row address $\text{rowAddr}[r-1:0]$ selects a many-word, while from the selected many-word, the column address $\text{colAddr}[q-1:0]$ selects the word addressed by the address $\text{addr}[p-1:0]$. Playing with the values of r and q an appropriate lay-out of the memory array can be designed.

In Figure 3.14 the block schematic for the resulting memory is presented. The second decoder – COLUMN DECODE – selects from the s m -bit words provided by the s COLUMN BLOCKs the word addressed by $\text{addr}[p-1:0]$.

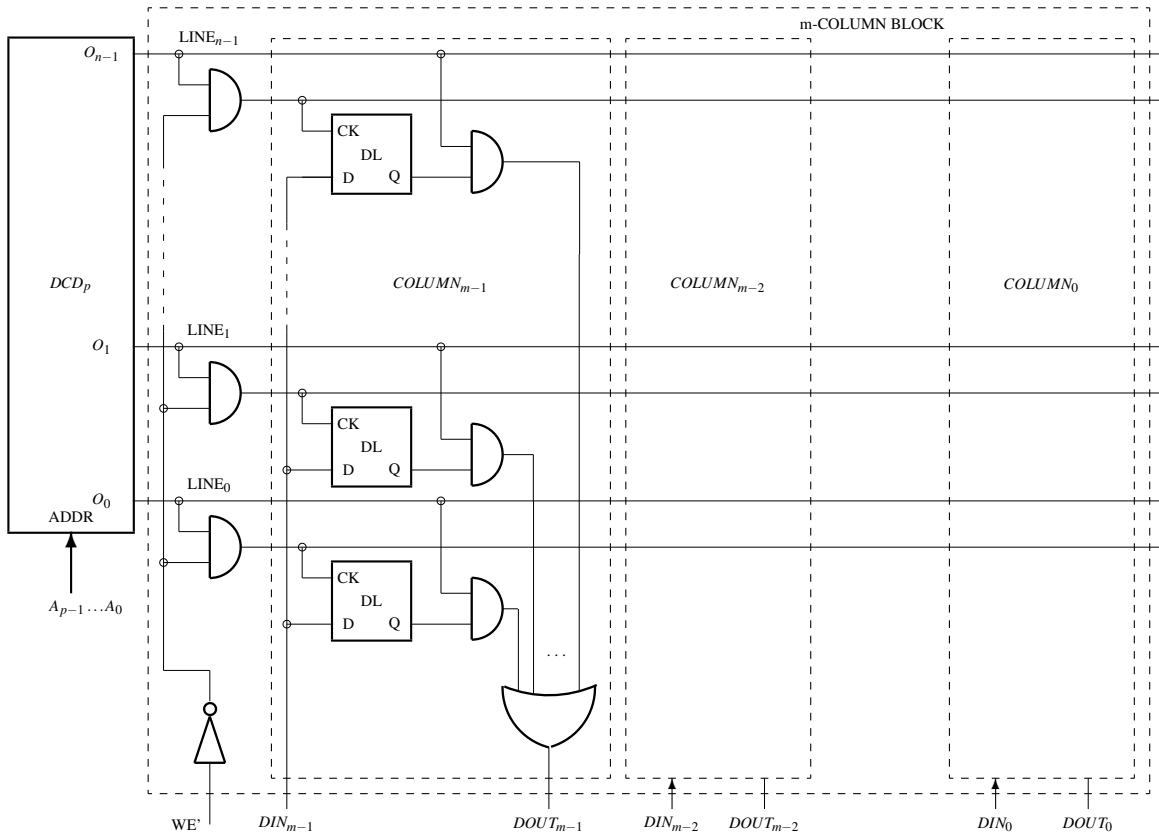


Figure 3.13: **The asynchronous m -bit word RAM.** Expanding the number of bits per word means to connect in parallel one-bit word memories which share the same decoder. Each COLUMN contains the storing latches and the AND-OR circuits for one bit.

While the size decoder for a one block memory version is in the same order with the number of words ($S_{DCD_p} \in 2^p$), the sum of the sizes of the two decoders in the two dimension version is much smaller, because usually $2^p \gg 2^r + 2^q$, for $p = r + q$. Thus, the area of the memory circuit is dominated only by the storage elements.

The second level of selection is based also on a shared decoder – COLUMN DECODER. It forms, with the s two-input ANDs a $DMUX_q$ – the **q-input DMUX** in Figure 3.14 – which distributes the write enable signals, we , to the selected m -column block. The same decoder is shared by the m s -input MUXs used to select the output word from the many-word selected by ROW DECODE.

The well known principle of "divide et impera" (*divide and conquer*) is applied when the address is divided in two parts, one for selecting a row and another for selecting a column. The access circuits is thus minimized.

Unfortunately, RAM has not the *function of memorizing*. It is only a storage support. Indeed, if we want to "memorize" the number 13, for example, we must store it to the address 131313, for example, and to keep in mind (to memorize) the value 131313, the place where the number is stored. And than,

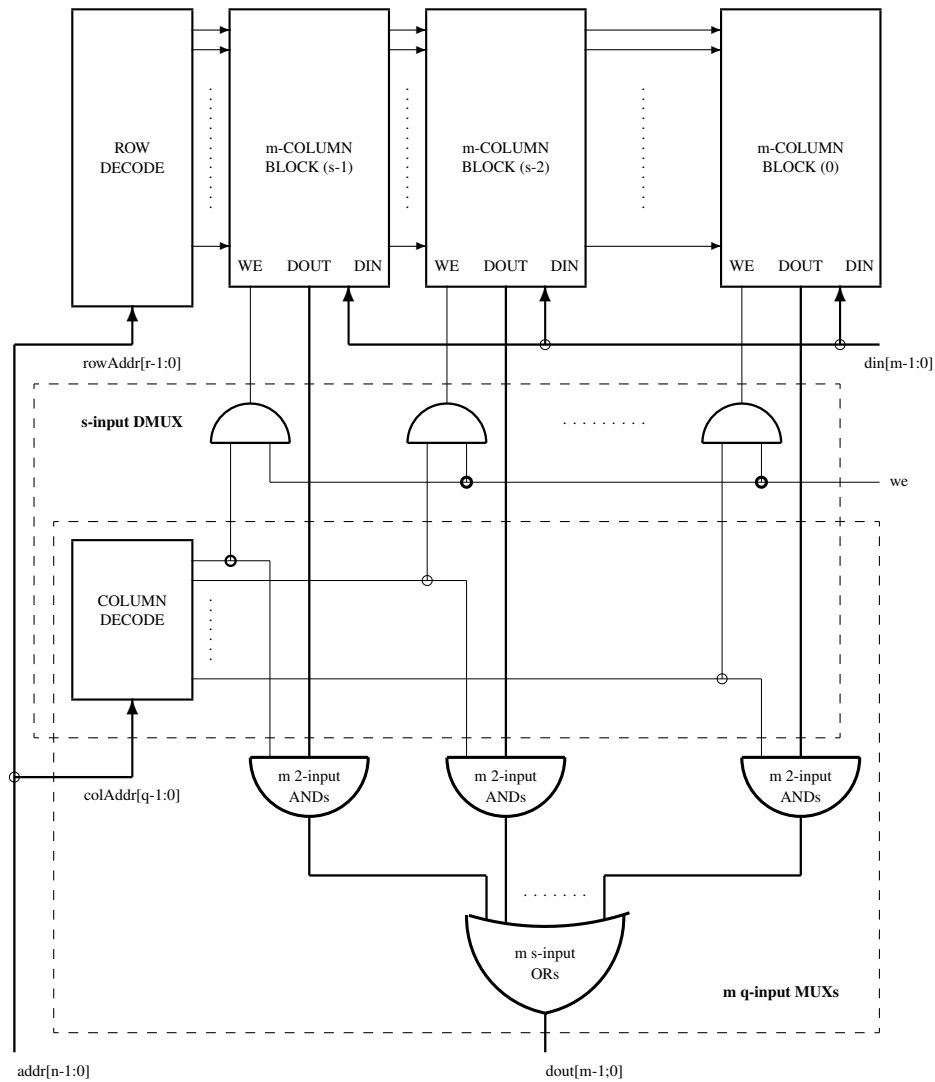


Figure 3.14: **RAM version with two dimension storage array.** A number of m -bit blocks are parallel connected and driven by the same row decoder. The column decoder selects to output an m -bit word from the $(s \times m)$ -bit row.

what's the help provided us by a the famous RAM memory? No one. Because RAM is not a memory, it becomes a memory only if the associated processor runs an appropriate procedure which allows us to forget about the address 131313. Another solution is provided by additional circuits used to improve the functionality (see the subsection about *Associative Memories*.)

3.5 The Serial-Parallel Composition: the Register

The last composition in 1-OS is the *serial-parallel composition*. The most representative circuit of this class is the *register*. The main application of register is to support the synchronous processes in a digital system. There are two typical use of the register:

- provides the *pipeline* connection between subsystems (see the subsections 2.5.1 *Pipelined connections*, and 3.3.2 *Pipeline structures*).
- stores the internal state of an automata (see the next chapter); the register is used to close of the second loop in a digital system.

Unlike the parallel compositions that *store asynchronously*, the circuits resulting from the serial-parallel compositions *store synchronously* the value applied on their inputs. The parallel compositions are used for designing *memory* systems, instead of the serial-parallel compositions, used to support the designing of the *control* structures in a digital system.

The skeleton of any contemporary digital design is based on registers, used to store, synchronously with the system clock, the overall state of the system. The Verilog (or VHDL) description of a structured digital design starts by defining the registers, and provides, usually, an *Register Transfer Logic* (RTL) description. An RTL code describe a set of registers interconnected through (simple uniform or complex random) combinational blocks. For a register is a non-transparent structure any loop configurations are supported. Therefore, the design is freed by the care of the unstable loops.

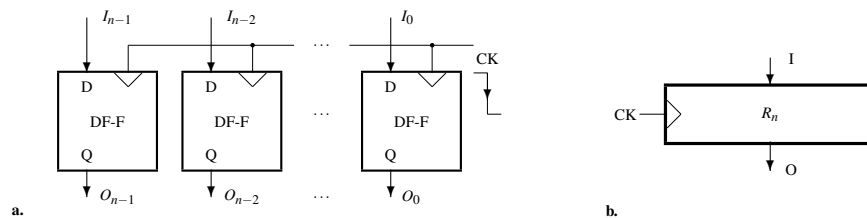


Figure 3.15: **The n -bit register.** **a.** The structure: a bunch of DF-F connected in parallel. **b.** The logic symbol.

Definition 3.3 An n -bit register, R_n , is made by parallel connecting a R_{n-1} with a D (master-slave) flip-flop. R_1 is a D flip-flop. \diamond

The register R_n , represented in Figure 3.15, is a *serial-parallel composition* in 1-OS because its elementary component, the D flip-flops, are serial compositions in 1-OS. Another possible definition is to build the register by serially connecting two n -bit latches. We know that the n -bit latch is a parallel extension in 1-OS. The clock must be applied to the two n -bit latches avoiding the simultaneous transparency.

VeriSim 3.11 An 8-bit enabled and resettable register with 2 unit time delay is described by the following Verilog module:

```

module register #(parameter n = 8)(output reg [n-1:0] out
                                ,
                                input [n-1:0] in
                                ,
                                input reset, enable, clock) ;

    always @(posedge clock) #2 if (reset) out <= 0 ;
                                else if (enable) out <= in ;
                                else out <= out ;

endmodule

```

The time behavior specified by #2 is added only for simulation purpose. The synthesizable version must avoid this unsynthesizable representation. \diamond

The main feature of the register assures its non-transparency, excepting an "undecided transparency" during a short time interval, $t_{SU} + t_H$, centered on the active edge of the clock signal. Thus, a new loop can be closed carelessly over a structure containing a register. Due to its non-transparency the register will be properly loaded with any value, even with a value depending on its own current content. This last feature is the main condition to close the loop of a synchronous automata - the structure presented in the next chapter.

3.6 Applications

Composing basic memory circuits with combinational structures result typical system configurations or typical functions to be used in structuring digital machines. The *pipeline* connection, for example, is a system configuration for speeding up a digital system using a sort of parallelism. This mechanism is already described in the subsections 2.5.1 *Pipelined connections*, and 3.3.2 *Pipeline structures*. Few other applications of the circuits belonging to 1-OS are described in this section. The first is a frequent application of 1-OS: the synchronous memory, obtained adding clock triggered structures to an asynchronous memory. The next is the *file register* – a typical storage subsystem used in the kernel of the almost all computational structures. The basic building block in one of the most popular digital device, the *Field Programmable Gate Array*, is also SRAM based structure. Follows the *content addressable memory* which is a hardware mechanism useful in controlling complex digital systems or for designing **genuine memory structures**: the *associative memories*.

3.6.1 Synchronous RAM

It is very hard to consider the time restriction imposed by the wave forms presented in Figure 3.12 when the system is requested to work at high speed. The system designer will be more comfortable with a memory circuit having all the time restrictions defined related *only* to the active edge of the system clock. The synchronous RAM (SRAM) is conceived to have all time relations defined related to the active edge of the clock signal. SRAM is the preferred embodiment of a storage circuit in the contemporary designs. It performs write and read operations synchronized with the active edge of the clock signal (see Figure 3.16).

VeriSim 3.12 The functional description of a synchronous RAM (0.5K of 64-bit words) follows:

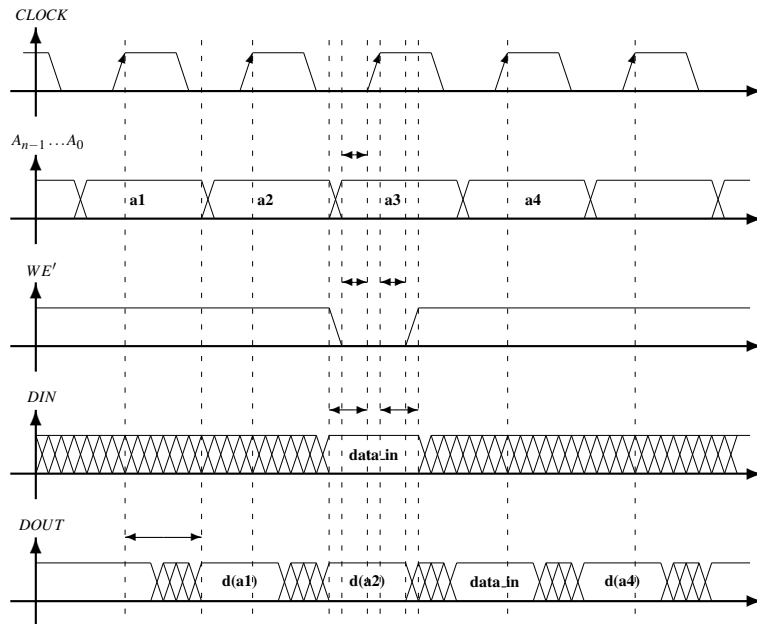


Figure 3.16: **Read and write cycles for SRAM.** For the *flow-through* version of a *SRAM* the time behavior is similar to a register. The set-up and hold time are defined related to the active edge of clock for all the input connections: *data*, *write-enable*, and *address*. The data output is also related to the same edge.

```

/* *****
File name:      sram.v
Circuit name:   Synchronous RAM
Description:    behavioral description of a synchronous RAM
***** */
module sram(    input    [63:0]  din  ,
               input    [8:0]   addr ,
               output reg [63:0]  dout ,
               input          we, clk);
    reg [63:0] mem[511:0];
    always @(posedge clk) if (we) dout <= din      ;
                          else  dout <= mem[addr]  ; // reading
    always @(posedge clk) if (we) mem[addr] <= din ; // writing
endmodule

```

◇

The previously described SRAM is the *flow-through* version of a SRAM. A pipelined version is also possible. It introduces another clock cycle delay for the output data.

3.6.2 Register File

The most accessible data in a computational system is stored in a small and fast memory whose locations are usually called **machine registers** or simply *registers*. In most usual embodiment they have actually the physical structure of a register. The machine registers of a computational (processing) element are organized in what is called *register file*. Because computation supposes two operands and one result in most of cases, two read ports and one write port are currently provided to the small memory used as register file (see Figure 3.17).

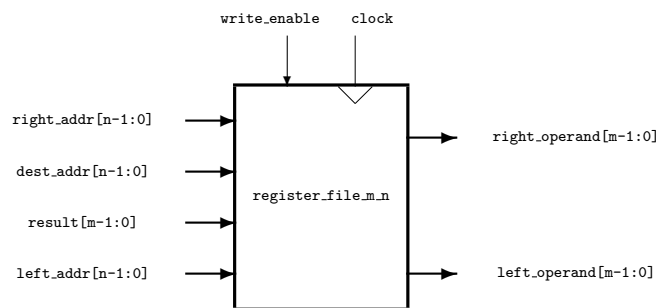


Figure 3.17: **Register file.** In this example it contains 2^n m -bit registers. In each clock cycle any two registers can be read and writing can be performed in anyone.

VeriSim 3.13 Follows the Verilog description of a register file containing 32 32-bit registers. In each clock cycle any two pair of registers can be accessed to be used as operands and a result can be stored in any one register.

```

/* *****
File name:      register_file.v
Circuit name:
Description:
***** */
module register_file(  output [31:0] left_operand    ,
                      output [31:0] right_operand   ,
                      input  [31:0] result          ,
                      input  [4:0]  left_addr       ,
                      input  [4:0]  right_addr      ,
                      input  [4:0]  dest_addr       ,
                      input                write_enable ,
                      input                clock      );
  reg [31:0] file [0:31];
  assign left_operand  = file[left_addr]  ,
         right_operand = file[right_addr] ;
  always @(posedge clock) if (write_enable) file[dest_addr] <= result;
endmodule

```

◇

The internal structure of a register file can be optimized using $m \times 2^n$ 1-bit clocked latches to store data and $2m$ -bit clocked latches to implement the master-slave mechanism.

3.7 Concluding About Memory Circuits

For the first time, in this chapter, both composition and loop are used to construct digital systems. The loop adds a new feature and the composition expands it. The chapter introduced only the basic concepts and the main ways to use them in implementing actual digital systems.

The first closed loop in digital circuits latches events Closing properly simple loops in small combinational circuits very useful effects are obtained. The most useful is the “latch effect” allowing to store certain temporal events. An internal loop is able to determine an **internal state** of the circuit which is independent in some extent from the input signals (the circuit controls a part of its inputs using its own outputs). Associating different internal states to different input events the circuit is able to **store** the input event in its internal states. The first loop introduces the first degree of **autonomy** in a digital system: *the autonomy of the internal state*. The resulting basic circuit for building memory systems is the *elementary latch*.

Meaningful circuits occur by composing latches The elementary latches are composed in different modes to obtain the main memory systems. The *serial composition* generates the **master-slave** flip-flop which is triggered by the *active edge* of the clock signal. The *parallel composition* introduces the concept of **random access memory**. The *serial-parallel composition* defines the concept of **register**.

Distinguishing between “how?” and “when?” At the level of the first order systems occurs a very special signal called **clock**. The clock signal becomes responsible for the *history sensitive processes* in a digital system. Each “clocked” system has inputs receiving information about “how” to switch and another special input – the clock input acting on one of its edge called the *active edge* of clock – and another special input indicating “when” the system switches. We call this kind of digital systems *synchronous systems*, because any change inside the system is triggered synchronously by the same edge (positive or negative) of the clock signal.

Registers and RAMs are basic structures First order systems provide few of the most important type of digital circuits used to support the future developments when new loops will be closed. The **register** is a synchronous subsystem which, because of its non-transparency, allows closing the next loop leading to the second order digital systems. Registers are used also for accelerating the processing by designing pipelined systems. The **random access memory** will be used as storage element in developing systems for processing a big amount of data or systems performing very complex computations. Both, data and programs are stored in RAMs.

RAM is not a memory, it is only a physical support Unfortunately RAM has not the function of memorizing. It is only a storage element. Indeed, when the word W is stored at the address A *we must memorize* the address A in order to be able to retrieve the word W . Thus, instead of memorizing W we must memorize A , or, as usual, we must have a mechanism to regenerate the address A . In conjunction with other circuits RAM can be used to build systems having the function of memorizing. Any memory system contains a RAM but not only a RAM, because memorizing means more than storing.

Memorizing means to associate Memorizing means both to store data and to retrieve it. The most “natural” way to design a memory system is to provide a mechanism able to associate the stored data with its location. In an associative memory to read means to find, and to write means to find a free location. The **associative memory** is the most perfect way of designing a memory, even if it is not always the most optimal as area (price), time and power.

To solve ambiguities a new loop is needed At the level of the first order systems the second latch problem can not be solved. The system must be more “intelligent” to solve the ambiguity of receiving synchronously contradictory commands. The system must know more about itself in order to be “able” to behave under ambiguous circumstances. Only a new loop will help the system to behave coherently. The next chapter, dealing with the second level of loops, will offer a robust solution to the second latch problem.

The storing and memory functions, typical for the first order systems, are not true computational features. We will see that they are only useful ingredients allowing to make digital computational systems efficient.

3.8 Problems

Stable/unstable loops

Problem 3.1 *Simulate in Verilog the unstable circuit described in **Example 3.1**. Use 2 unit time (#2) delay for each circuit and measure the frequency of the output signal.*

Problem 3.2 Draw the circuits described by the following expressions and analyze their stability taking into account all the possible combinations applied on their inputs:

$$d = b(ad)' + c$$

$$d = (b(ad)' + c)'$$

$$c = (ac' + bc)'$$

$$c = (a \oplus c) \oplus b.$$

Simple latches

Problem 3.3 Illustrate the second latch problem with a Verilog simulation. Use also versions of the elementary latch with the two gates having distinct propagation times.

Problem 3.4 Design and simulate an elementary clocked latch using a NOR latch as elementary latch.

Problem 3.5 Let be the circuit from Figure 3.18. Indicate the functionality and explain it.

Hint: emphasize the structure of an elementary multiplexer.

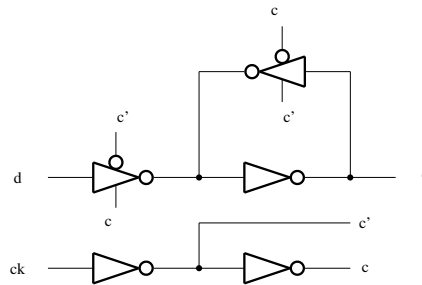


Figure 3.18: ?

Problem 3.6 Explain how it works and find an application for the circuit represented in Figure 3.19.

Hint: Imagine the tristate drivers are parts of two big multiplexors.

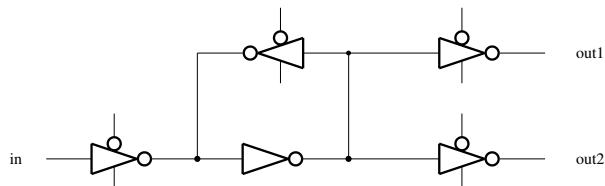


Figure 3.19: ?

Master-slave flip-flops

Problem 3.7 Design an asynchronously presettable master-slave flip-flop.

Hint: to the slave latch must be added asynchronous set and reset inputs (S' and R' in the NAND latch version, or S and R in the NOR latch version).

Problem 3.8 Design and simulate in Verilog a positive edge triggered master-slave structure.

Problem 3.9 Design a positive edge triggered master slave structure without the clock inverter.

Hint: use an appropriate combination of latches, one transparent on the low level of the clock and another transparent on the high level of the clock.

Problem 3.10 Design the simulation environment for illustrating the master-slave principle with emphasis on the set-up time and the hold time.

Problem 3.11 Let be the circuit from Figure 3.20. Indicate the functionality and explain it. Modify the circuit to be triggered by the other edge of the clock.

Hint: emphasize the structures of two clocked latches and explain how they interact.

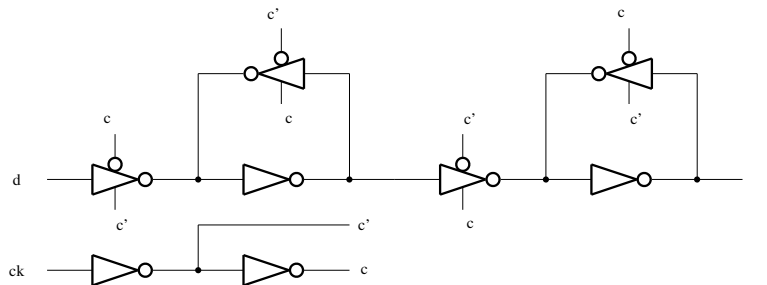


Figure 3.20: ?

Problem 3.12 Let be the circuit from Figure 3.21. Indicate the functionality and explain it. Assign a name for the questioned input. What happens if the NANDs are substituted with NORs. Rename the questioned input. Combine both functionality designing a more complex structure.

Hint: go back to Figure 3.3c.

Enabled circuits

Problem 3.13 An n -bit latch stores the n -bit value applied on its inputs. It is transparent on the low level of the clock. Design an enabled n -bit latch which stores only in the clock cycle in which the enable input, en , take the value 1 synchronized with the positive edge of the clock. Define the set-up time and the hold time related to the appropriate clock edge for data input and for the enable signal.

Problem 3.14 Provide a recursive Verilog description for an n -bit enabled latch.

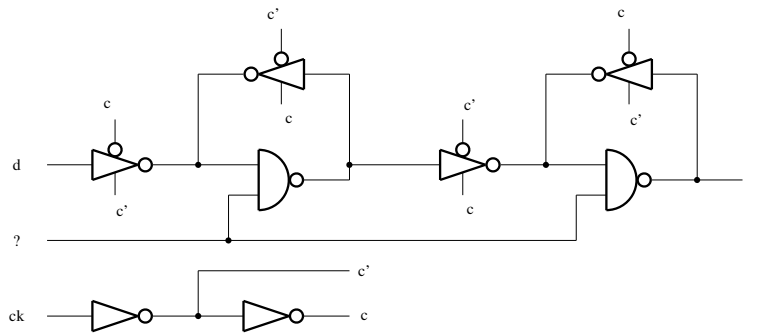


Figure 3.21: ?

RAMs

Problem 3.15 Explain the reason for t_{ASU} and for t_{AH} in terms of the combinational hazard.

Problem 3.16 Explain the reason for t_{DSU} and for t_{DH} .

Problem 3.17 Provide a structural description of the RAM circuit represented in Figure 3.11 for $m = 256$. Compute the size of the circuit emphasizing both the weight of storing circuits and the weight of the access circuits.

Problem 3.18 Design a 256-bit RAM using a two-dimensional array of 16×16 latches in order to balance the weight of the storing circuits with the weight of the accessing circuits.

Problem 3.19 Design the flow-through version of SRAM defined in Figure 3.16.

Hint: use additional storage circuits for address and input data, and relate the WE' signal with the clock signal.

Problem 3.20 Design the register to latch version of SRAM defined in Figure 3.22.

Hint: the write process is identical with the flow-through version.

Problem 3.21 Design the pipeline version of SRAM defined in Figure 3.22.

Hint: only the output storage device must be adapted.

Registers

Problem 3.22 Provide a recursive description of an n -bit register. Prove that the (algorithmic) complexity of the concept of register is in $O(n)$ and the complexity of a certain register is in $O(\log n)$.

Problem 3.23 Draw the schematic for an 8-bit enabled and resettable register. Provide the Verilog environment for testing the resulting circuit. Main restriction: the clock signal must be applied only directly to each D flip-flop.

Hint: an enabled device performs its function only if the enable signal is active; to reset a register means to load it with the value 0.

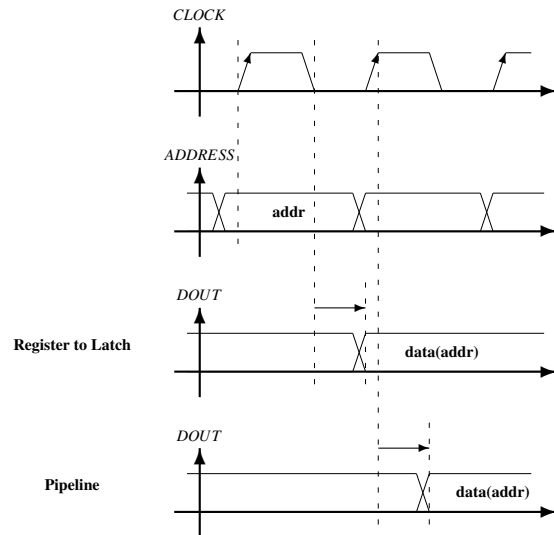


Figure 3.22: **Read cycles.** Read cycle for the *register to latch* version and for the *pipeline* version of SRAM.

Problem 3.24 Add to the register designed in the previous problem the following feature: the content of the register is shifted one binary position right (the content is divided by two neglecting the remainder) and on most significant bit (MSB) position is loaded the value of the one input bit called SI (serial input). The resulting circuit will be commanded with a 2-bit code having the following meanings:

nop : the content of the register remains unchanged (the circuit is disabled)

reset : the content of the register becomes zero

load : the register takes the value applied on its data inputs

shift : the content of the register is shifted.

Problem 3.25 Design a serial-parallel register which shifts 16 16-bit numbers.

Definition 3.4 The serial-parallel register, $SPR_{n \times m}$, is made by a $SPR_{(n-1) \times m}$ serial connected with a R_m . The $SPR_{1 \times m}$ is R_m . \diamond

Hint: the serial-parallel register, $SPR_{n \times m}$ can be seen in two manners. $SPR_{n \times m}$ consists in m parallel connected serial registers SR_n , or $SPR_{n \times m}$ consists in n serially connected registers R_m . We prefer usually the second approach. In Figure 3.23 is shown the serial-parallel $SPR_{n \times m}$.

Problem 3.26 Let be t_{SU} , t_H , t_p , for a register and t_{pCLC} the propagation time associated with the CLC loop connected with the register. The maximal and minimal value of each is provided. Write the relations governing these time intervals which must be fulfilled for a proper functioning of the loop.

Pipeline systems

Problem 3.27 Explain what is wrong in the following always construct used to describe a pipelined system.

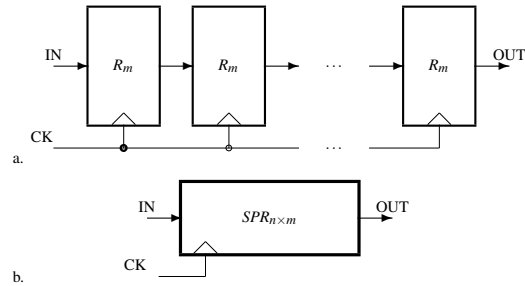


Figure 3.23: **The serial-parallel register.** **a.** The structure. **b.** The logic symbol.

```

module pipeline      #(parameter  n = 8, m = 16, p = 20)
                      (output    reg[m-1:]  output_reg ,
                      input     wire[n-1:0] in ,
                      clock );

  reg[n-1:0]         input_reg ;
  reg[p-1:0]         pipeline_reg ;
  wire[p-1:0]        out1 ;
  wire[m-1:0]        out2 ;
  clc1    first_clc(out1 , input_reg );
  clc2    second_clc(out2 , pipeline_reg );

  always @(posedge clock) begin    input_reg = in ;
                                   pipeline_reg = out1 ;
                                   output_reg = out2 ;
  end

endmodule
module clc1(out1 , in1);
  // ...
endmodule
module clc2(out2 , in2);
  // ...
endmodule

```

Hint: revisit the explanation about blocking and nonblocking evaluation in Verilog.

Register file

Problem 3.28 Draw register_file_16.4 at the level of registers, multiplexors and decoders.

Problem 3.29 Evaluate for register_file_32.5 minimum input arrival time before clock (t_{in_reg}), minimum period of clock (T_{min}), maximum combinational path delay (t_{in_out}) and maximum output required time after clock (t_{reg_out}) using circuit timing from Appendix Standard cell libraries.


```

module memory #(parameter n=32, m=10)
  (  output reg [n-1:0] dataOut      , // data output
    input      [n-1:0] dataIn        , // data input
    input      [m-1:0] readAddr      , // read address
    input      [m-1:0] writeAddr     , // write address
    input      we                    , // write enable
    input      enable                , // module enable
    input      clock                 );

  reg [n-1:0] memory[0:(1 << m)-1];

  always @(posedge clock) if (enable) begin
    if (we) memory[writeAddr] <= dataIn ;
    dataOut <= memory[readAddr]         ;
  end

endmodule

```

Design the module bufferMemory.

Project 3.2 *Design a systolic system for multiplying a band matrix of maximum width 16 with a vector. The operands are stored in serial registers.*

Chapter 4

AUTOMATA: Second order, 2-loop digital systems

*The Tao of heaven is impartial.
If you perpetuate it, it perpetuates you.*

Lao Tzu¹

*Perpetuating the inner behavior is the
magic of the second loop.*

The next step in building digital systems is to add a new loop over systems containing 1-OS. This new loop must be introduced carefully so as the system remains *stable* and *controllable*. One of the most reliable ways is to build synchronous structures, that means to close the loop through a way containing a register. The non-transparency of registers allows us to separate with great accuracy the current state of the machine from the next state of the same machine.

This second loop increases the autonomous behavior of the system including it. As we shall see, in 2-OS each system has the autonomy of *evolving* in the state space, partially independent from the input dynamics, rather than in 1-OS in which the system has only the autonomy of preserving a certain state.

The basic structure in 2-OS is the *automaton*, a digital system with outputs evolving according to two variables: the input variable and a “hidden” internal variable named the *internal state variable*, simply the em state. The autonomy is given by the internal effect of the state. The behavior of the circuit output can not be explained only by the evolution of the input, the circuit has an internal autonomous evolution that “memorizes” previous events. Thus the response of the circuit to the actual input takes into account the more or less recent history. The *state space* is the space of the internal state and its dimension is responsible for the behavioral complexity. Thus, the degree of autonomy depends on the dimension of the state space.

An automaton is built closing a loop over a 1-OS represented by a collection of latches. The loop can be structured using the previous two type of systems. Thus, there are two type of automata:

- *asynchronous automata*, for which the loop is closed over **unlocked latches**, through combinational circuit and/or **unlocked latches** as in Figure 4.1a

¹Quote from *Tao Te King* of Lao Tzu translated by Brian Browne Walker.

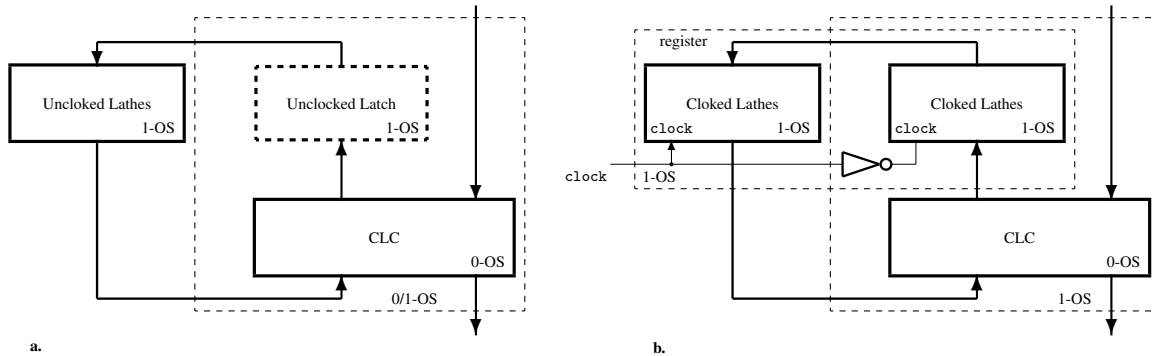


Figure 4.1: **The two type of 2-OS.** **a.** The asynchronous automata with a hazardous loop over a transparent latch. **b.** The synchronous automata with a edge clock controlled loop closed over a non-transparent register.

- *synchronous automata*, having the loop closed through an 1-OS and all latches are clocked latches connected on the loop in master-slave configurations (see Figure 4.1b).

Our approach will be focused on the synchronous automata, after considering only in the first subchapter an asynchronous automaton used to optimize the internal structure of the widely used flip-flop: DFF.

4.1 Basic definitions in automata theory

Definition 4.1 An automaton, A , is defined by the following 5-uple:

$$A = (X, Y, Q, f, g)$$

where:

X : the finite set of input variables

Y : the finite set of output variables

Q : the set of state variables

f : the state transition function, described by $f : X \times Q \rightarrow Q$

g : the output transition function, with one of the following definitions:

- $g : X \times Q \rightarrow Y$ for Mealy type automaton
- $g : Q \rightarrow Y$ for Moore type automaton
- $g(q) = q$ for $Y \equiv Q$, where $q \in Q$ for half-automaton, symbolized with $A_{1/2}$.

At each clock cycle the state of the automaton switches and the output takes the value according to the new state (and the current input, in Mealy's approach). \diamond

Definition 4.2 A finite automaton, FA, is an automaton with Q a finite set. \diamond

FA is a complex circuit because the size of its definition depends by $|Q|$.

Definition 4.3 A recursively defined ***n*-state automaton**, *n*-SA, is an automaton with $|Q| \in O(f(n))$. \diamond

An *n*-SA has a finite (usually short) definition depending by one or many parameters. Its size will depend by parameters. Therefore, it is a simple circuit.

Definition 4.4 An **initial state** is a state having no predecessor state. \diamond

Definition 4.5 An **initial automaton** is an automaton having a set of initial states, Q' , which is a subset of Q , $Q' \subset Q$. \diamond

Definition 4.6 A **strict initial automaton** is an automaton having only one initial state, $Q' = \{q_0\}$. \diamond

A strict initial automaton is defined by:

$$A = (X, Y, Q, f, g; q_0)$$

and has a special input, called **reset**, used to led the automaton in the initial state q_0 . If the automaton is initial only, the input reset switches the automaton in one, specially selected, initial state.

Definition 4.7 The delayed (*Mealy or Moore*) automaton is an automaton with the output values generated through a (delay) register, thus the current output value corresponds to the previous internal state of the automaton, instead of the current value of the state, as in non-delayed version. \diamond

The half automaton is an automaton with identity function as the output function (see Figure 4.2a,b) defined for two reasons:

- many optimization techniques are related only with the loop circuits of the automaton. The main feature of an automaton is the autonomy and the associated half-automaton, concept which describes especially this type of behavior
- there are applications that use directly the state as outputs.

All kind of automata can be described starting from a half-automaton, adding only combinational (no loops) circuits and/or memory (one loop) circuits. In Figure 4.2 are presented all the four types of automata:

Mealy automaton : results connecting to the “output” of an $A_{1/2}$ the output CLC that receives also the input X (Figure 4.2c) and computes the output function g ; a combinational way occurs between the input and the output of this automaton allowing a fast response, in the same clock cycle, to the input variation

Moore automaton : results connecting to the “output” of an $A_{1/2}$ the output CLC (Figure 4.2d) that computes the output function g ; this automaton reacts to the input signal in the next clock cycle

delayed Mealy automaton : results serially connecting a register, R, to the output of the Mealy automaton (Figure 4.2e); this automaton reacts also to the input signal in the next clock cycle, but the output is hazard free because it is registered

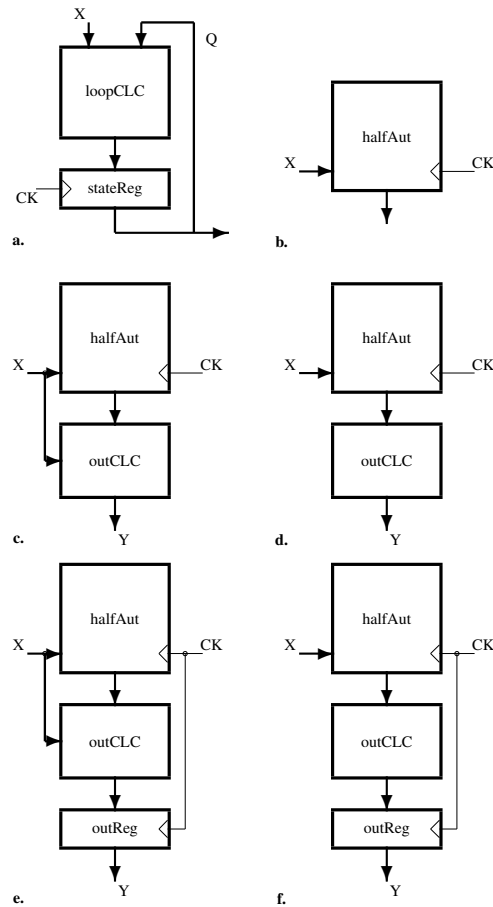


Figure 4.2: **Automata types.** **a.** The structure of the half-automaton ($A_{1/2}$), the no-output automaton: the state is generated by the previous state and the previous input. **b.** The logic symbol of half-automaton. **c.** Immediate Mealy automaton: the output is generated by the current state and the current input. **d.** Immediate Moore automaton: the output is generated by the current state. **e.** Delayed Mealy automaton: the output is generated by the previous state and the previous input. **f.** Delayed Moore automaton: the output is generated by the previous state.

delayed Moore automaton : results serially connecting a register, R , to the output of the Moore automaton (Figure 4.2f); this automaton reacts to the input signal with a two clock cycles delay.

Real applications use all the previous type of automata, because they react with different delay to the input change. The registered outputs are preferred if possible.

Theorem 4.1 *The time relation between the input value and the output value is the following for the four types of automata:*

1. for Mealy automaton the output to the moment t , $y(t) \in Y$ depends on the current input value, $x(t) \in X$, and by the current state, $q(t) \in Q$, i.e., $y(t) = g(x(t), q(t))$
2. for delayed Mealy automaton and Moore automaton the output corresponds with the input value from the previous clock cycle:

- $y(t) = g(x(t-1), q(t-1))$ for Mealy delayed automaton
- $y(t) = g(q(t)) = g(f(x(t-1), q(t-1)))$ for Moore automaton

3. for delayed Moore automaton the input transition acts on the output transition delayed with two clock cycles:

$$y(t) = g(q(t-1)) = g(f(x(t-2), q(t-2))). \diamond$$

Proof The proof is evident starting from the previous two definitions. \diamond

The possibility emphasized by this theorem is that we dispose of automata with different time reaction to the input variations. The Mealy automaton follows immediate the input transitions, delayed Mealy and Moore automata react with one clock cycle delay to the input transitions and delayed Moore automaton delays with two cycles the response to the input.

The symbols from the sets X , Y , and Q are binary coded using bits specified by X_0, X_1, \dots for X , Y_0, Y_1, \dots for Y , Q_0, Q_1, \dots for Q .

Actually, all implementable automata are finite. Traditionally, the term *finite automaton* is used to distinguish a subset of automata whose behavior is described using a constant number of states. Even if the input string is *infinite*, the behavior of the automaton is limited to a trajectory traversing a constant (*finite*) number of states. A finite automaton will be an automaton having a random combinational function for its transition functions f and g . Therefore, a finite automaton is a complex structure.

A “non-finite” automaton that is an automaton designed to evolve in a state space proportional with the length of the input string. Now, if the input string is “*infinite*” the number of states must be also “*infinite*”. Such an automaton can be defined only if its transition function is simple. Its combinational loop is a simple circuit even if it can be a big one. The “non-finite” automaton has a number of states that does not affect the definition (see the following examples of counters, for sum prefix automaton, ...). We classify the automata in two categories:

- “non-finite”, recursive defined, simple automata, called **functional automata**, or simply *automata*
- non-recursive defined, complex automata, called **finite automata**.

We continue this chapter with an example of asynchronous circuit, because of its utility and because we intend to show how complex is the management of its behavior. We will continue presenting only synchronous automata, starting with *small* automata having only two states (the smallest state space). We will continue with *simple*, recursive defined automata and we will end with finite automata, that are the most *complex* automata.

4.2 Finite Automata: the Complex Automata

After presenting the *elementary small automata* and the *large and simple functional automata* it is the time to discuss about the **complex automata**. The main property of these automata is to use a random combinational circuit, CLC, for computing the state transition function and the output transition function. Designing a finite automaton means mainly to design two CLC: the loop CLC (associated to the state transition function f) and the output CLC (associated to the output transition function g).

4.2.1 Representing finite automata

A finite automaton is represented by defining its transition functions f , the state transition function, and g , the output transition function. For a half-automaton only the function f defined.

Flow-charts

A flow-chart contains for each state a circle and for each type of transition an arrow. In each clock cycle the automaton “runs” on an arrow going from the current state to the next state. In our simple model the “race” on arrow is done in the moment of the active edge of the clock.

The flow-chart for a half-automaton The first version is a pure symbolic representation, where the flow chart is marked on each circle with the name of the state, and on each arrow with the transition condition, if any. The initial states can be additionally marked with the minus sign (-), and the final states can be additionally marked with the plus sign (+).

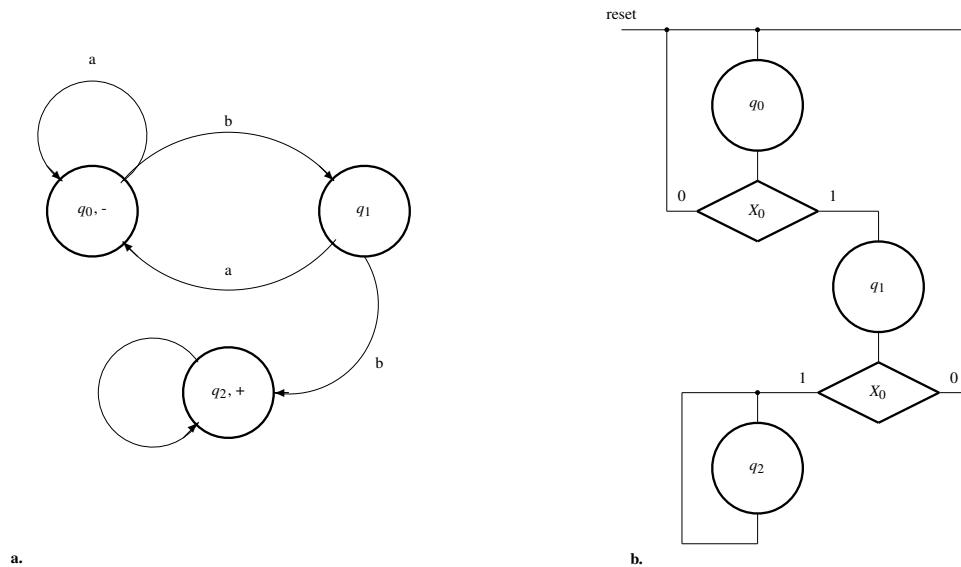


Figure 4.3: **Example of flow-chart for a half-automaton.** The machine is a “double b detector”. It stops when the first bb occurs.

The second version is used when the input are considered in the binary form. Instead of arches are used rhombuses containing the symbol denoting a binary variable.

Example 4.1 Let be a finite half-automaton that receives on its input strings containing symbols from the alphabet $X = \{a, b\}$. The machine stops in the final state when the first sequence bb is received. The first version of the associated flow-chart is in Figure 4.3a. Here is how the machine works:

- the initial state is q_0 ; if a is received the machine remains in the same state, else, if b is received, then the machine switch in the state q_1
- in the state q_1 the machine “knows” that one b was just received; if a is received the half-automaton switch back in q_0 , else, if b is received, then the machine switch in q_2
- q_2 is the final state; the next state is unconditionally q_2 .

The second version uses tests represented by a rhombus containing the tested binary input variable (see (Figure 4.3b). The input I takes the binary value 0 for the symbol a and the binary value 1 for the symbol b . \diamond

The second version is used mainly when a circuit implementation is envisaged.

The flow-chart for a Moore automaton When an automaton is represented the output behavior must be also included.

The first, pure symbolic version contains in each circle besides, the name of the state, the value of the output in that state. The output of the automaton shows something which is meaningful for the user. Each state generates an output value that can be different from the state's name. The output set of value are used to classify the state set. The input events are mapped into the state set, and the state set is mapped into the output set.

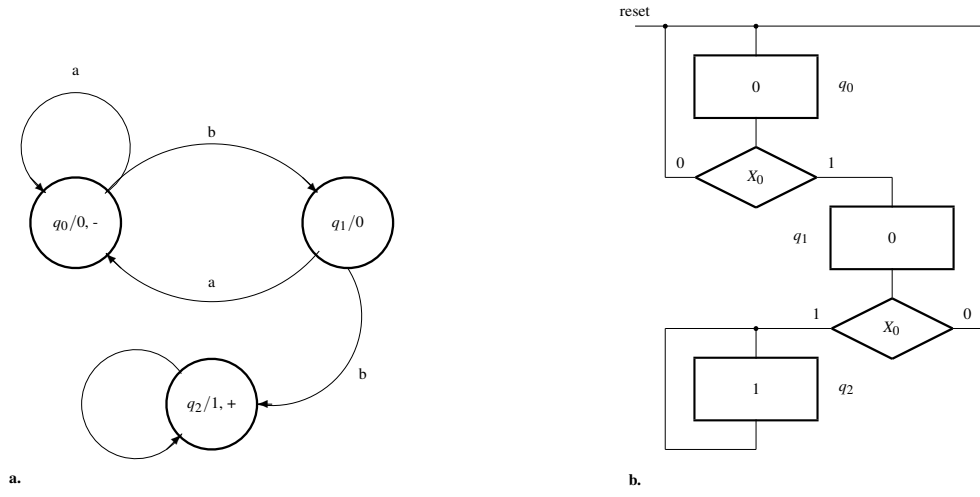


Figure 4.4: **Example of flow-chart for a Moore automaton.** The output of this automaton tells us: “ bb was already detected”.

The second uses for each pair state/output one rectangle. Inside of the rectangle is the value of the output and near to it is marked the state (by its name, by its binary code,, or both).

Example 4.2 The problem solved in the previous example is revisited using an automaton. The output set is $Y = \{0, 1\}$. If the output takes the value 1, then we learn that a double b was already received. The state set $Q = \{q_0, q_1, q_2\}$ is divided in two classes: $Q^0 = \{q_0, q_1\}$ and $Q^1 = \{q_2\}$. If the automaton stays in Q^0 with $out = 1$, then it is looking for bb . If the automaton stays in Q^1 with $out = 1$, then it stopped investigating the input because a double b was already received.

The associated flow-chart is in, in the first version represented by Figure 4.4a. The states q_0 and q_1 belong to Q^0 because in the corresponding circles we have $q_0/0$ and $q_1/0$. The state q_2 belongs to Q^1 because in the corresponding circle we have $q_2/1$. Because the evolution from q_2 does not depend by input, the arrow emerging from the corresponding circle is not labelled.

The second version (see Figure 4.4b) uses three rectangles, one for each state. \diamond

A meaningful event on the input of a Moore automaton is shown on the output with a delay of a clock cycle. All goes through the state set. In the previous example, if the second b from bb is applied on the input in the period T_i of the clock cycle, then the automaton points out the event in the period T_{i+1} of the clock cycle.

The flow-chart for a Mealy automaton The first, pure symbolic version contains on each arrow besides, the name of the condition, the value of the output generated in the state where the arrow starts with the input specified on the arrow.

The Mealy automaton reacts on its outputs more promptly to a meaningful input event. The output value depends on the input value from the same clock cycle.

The second, implementation oriented version uses rectangles to specify the output's behavior.

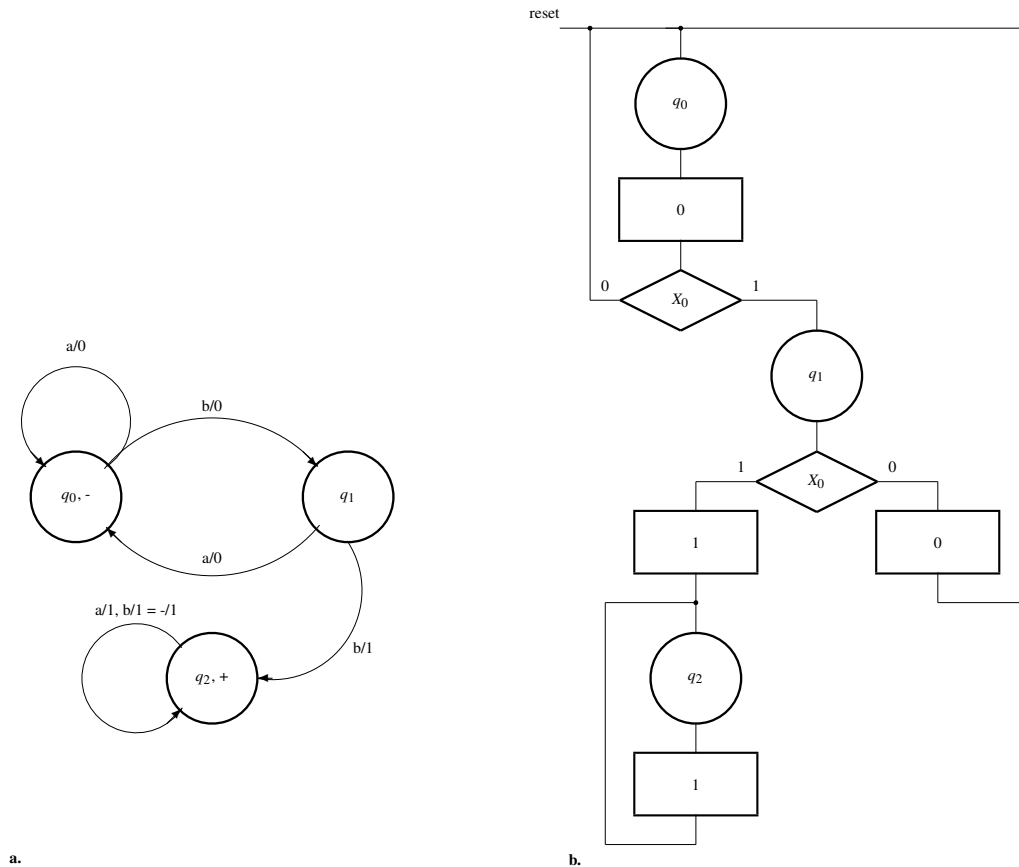


Figure 4.5: **Example of flow-chart for a Mealy automaton.** The occurrence of the second b from bb is detected as fast as possible.

Example 4.3 Let us solve again the same problem of bb detection using a Mealy automaton. The resulting flow-chart is in Figure 4.5a. Now the output is activated ($out = 1$) when the automaton is in the state q_1 (one b was detected in the previous cycle) and the input takes the value b . The same condition

triggers the switch in the state q_2 . In the final state q_2 the output is unconditionally 1. In the notation $-/1$ the sign $-$ stands for “don’t care”.

Figure 4.5b represents the second representation. \diamond

We can say the Mealy automaton is a “transparent” automaton, because a meaningful change on its inputs goes directly to its output.

Transition diagrams

Flow-charts are very good to offer an intuitive image about how automata behave. The concept is very well represented. But, automata are also actual machines. In order to help us to provide the real design we need different representation. Transition diagrams are less intuitive, but they work better for helping us to provide the image of the circuit performing the function of a certain automaton.

Transition diagrams uses Vetch-Karnaugh diagrams, VKD, for representing the transition functions. The representation maps the VKD describing the state set of the automaton into the VKDs defining the function f and the function g .

Transition diagrams are about real stuff. Therefore, the symbols like a, b, q_0, \dots must be coded binary, because a real machine work with bits, 0 and 1, not with symbols.

The output is already coded binary. For the input symbols the code is established by “the user” of the machine (similarly the output codes have been established by “the user”). Let say, for the input variable, X_0 , was decided the following codification: $a \rightarrow X_0 = 0$ and $b \rightarrow X_0 = 1$.

Because the actual value of the state is “hidden” from the user, the designer has the freedom to assign the binary values according to its own (engineering) criteria. Because the present approach is a theoretical one, we do not have engineering criteria. Therefore, we are completely free to assign the binary codes. Two option are presented:

option 1: $q_0 = 00, q_1 = 01, q_2 = 10$

option 2: $q_0 = 00, q_1 = 10, q_2 = 11$

For both the external behavior of the automaton must be the same.

Transition diagrams for half-automata The transition diagram maps the reference VKD into the next state VKD, thus defining the state transition function. Results a representation ready to be used to design and to optimize the physical structure of a finite half-automaton.

Example 4.4 The flow-chart from Figure 4.3 has two different correspondent representations as transition diagrams in Figure 4.6, one for the option 1 of coding (Figure 4.6a), and another for the option 2 (Figure 4.6b).

In VKD S_1, S_0 each box contains a 2-bit code. Three of them are used to code the states, and one will be ignored. VKD S_1^+, S_0^+ represents the transition from the corresponding states. Thus, for the first coding option:

- from the state coded 00 the automaton switch in the state 0x, that is to say:
 - if $X_0 = 0$ then the next state is 00 (q_0)
 - if $X_0 = 1$ then the next state is 01 (q_1)

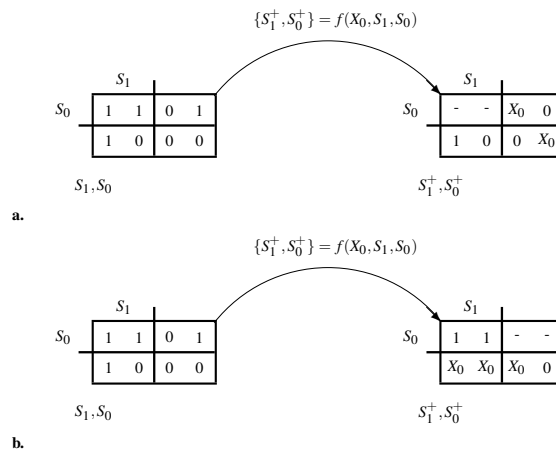


Figure 4.6: **Example of transition diagram for a half-automaton.** a. For the option 1 of coding. b. For the option 2 of coding.

- from the state coded 01 the automaton switches to the state X_0 , that is to say:
 - if $X_0 = 0$ then the next state is 00 (q_0)
 - if $X_0 = 1$ then the next state is 10 (q_2)
- from the state coded 10 the automaton switches to the same state, 10 that is the final state
- the transition from 11 is not defined.

If in the clock cycle T_i the state of the automaton is S_1, S_0 (defined in the reference VKD), then in the next clock cycle, T_{i+1} , the automaton switches to the state S_1^+, S_0^+ (defined in the next state VKD).

For the second coding option:

- from the state coded 00 the automaton switches to the state X_00 , that is to say:
 - if $X_0 = 0$ then the next state is 00 (q_0)
 - if $X_0 = 1$ then the next state is 10 (q_1)
- from the state coded 10 the automaton switches to the state X_0X_0 , that is to say:
 - if $X_0 = 0$ then the next state is 00 (q_0)
 - if $X_0 = 1$ then the next state is 11 (q_2)
- from the state coded 11 the automaton switches to the same state, 11 that is the final state
- the transition from 01 is not defined.

◇

The transition diagram can be used to extract the Boolean functions of the loop of the half-automaton.

Example 4.5 *The Boolean function of the half-automaton working as “double b detector” can be extracted from the transition diagram represented in Figure 4.6a (for the first coding option). Results:*

$$S_1^+ = S_1 + X_0S_0$$

$$S_0^+ = X_0S_1'S_0'$$

◇

Transition diagrams Moore automata The transition diagrams define the two transition functions of a finite automaton. To the VKDs describing the associated half-automaton is added another VKD describing the output’s behavior.

Example 4.6 *The flow-chart from Figure 4.4 have a correspondent representation in the transition diagrams from Figure 4.7a or Figure 4.7b. Besides the transition diagram for the state, the output transition diagrams are presented for the two coding options.*

For the first coding option:

- for the states coded with 00 and 01 the output has the value 0
- for the state coded with 10 the output has the value 1
- we do not care about how works the function g for the state coded with 11 because this code is not used in defining our automaton (the output value can 0 or 1 with no consequences on the automaton’s behavior).

◇

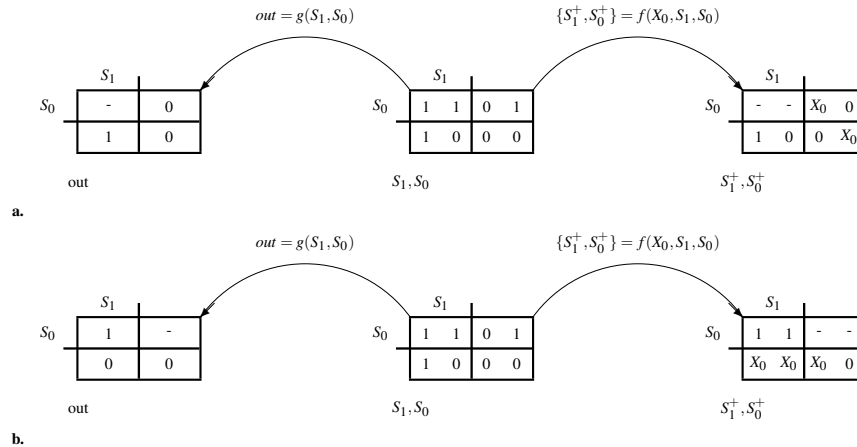


Figure 4.7: Example of transition diagram for a Moore automaton.

Example 4.7 *The resulting output function is:*

$$out = S_1.$$

Now the resulting automaton circuit can be physically implemented, in the version resulting from the first coding option, as a system containing a 2-bit register and few gates. Results the circuit in Figure 4.8, where:

- the 2-bit register is implemented using two resetable D flip-flops
- the combinational loop for state transition function consists in few simple gates
- the output transition function is so simple as no circuit are needed to implement it.

When `reset = 1` the two flip-flops switch in 0. When `reset = 0` the circuit starts to analyze the stream received on input symbol by symbol. In each clock cycle a new symbol is received and the automaton switches according to the new state computed by three gates. \diamond

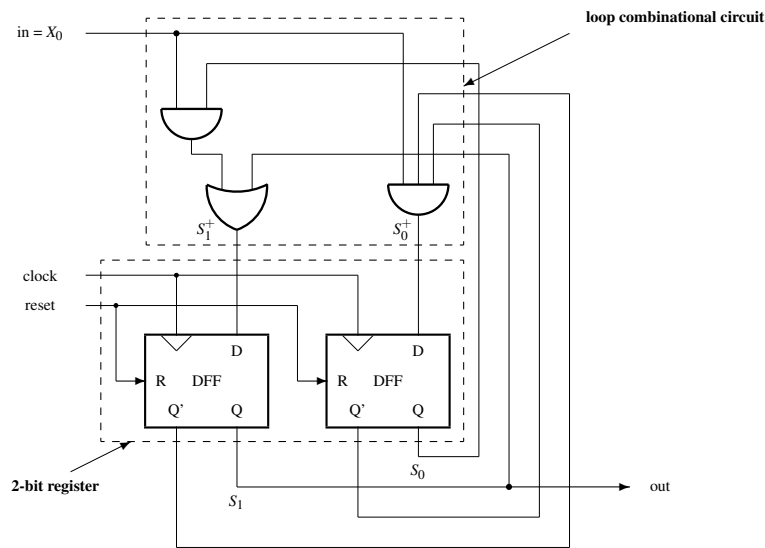


Figure 4.8: The Moore version of “bb detector” automaton.

Transition diagrams Mealy automata The transition diagrams for a Mealy automaton are a little different from those of Moore, because the output transition function depends also by the input variable. Therefore the VKD defining g contains, besides 0s and 1s, the input variable.

Example 4.8 Revisiting the same problem result, in Figure 4.9 the transition diagrams associated to the flow-chart from Figure 4.5.

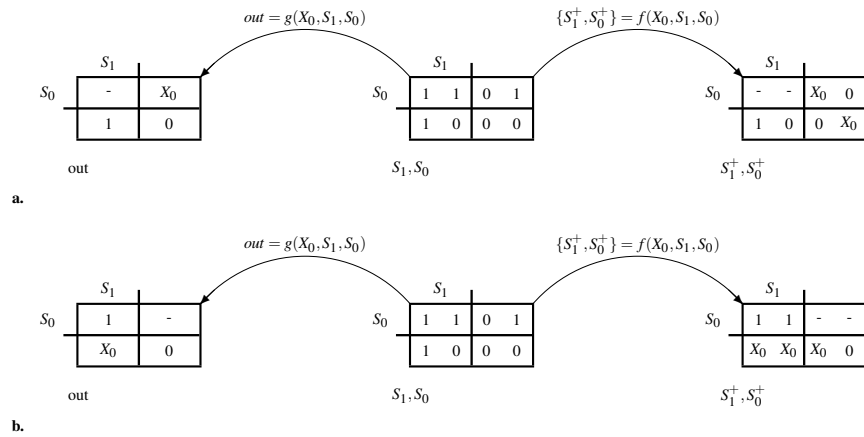


Figure 4.9: Example of transition diagram for a Mealy automaton.

The two functions f are the same. The function g is defined for the first coding option (Figure 4.9a) as follows:

- in the state coded by 00 (q_0) the output takes value 0
- in the state coded by 01 (q_1) the output takes value x
- in the state coded by 10 (q_2) the output takes value 1
- in the state coded by 11 (unused) the output takes the “don’t care” value

Extracting the function out results:

$$out = S_1 + X_0S_0$$

a more complex from compared with the Moore version. (But fortunately $out = S_1^+$, and the same circuits can be used to compute both functions. Please ignore. Engineering stuff.)

◇

Procedures

In the description and synthesis of finite automata, we will directly use the representations in HDL (in our case Verilog), avoiding the descriptions that use graphs, charts or Veitch-Karnaugh diagrams. We will start from the symbolic description of the sets X, Y, Q converted into a binary form (see defines.hv), and we will continue with the description of the associated semiautomaton (see halfAutomaton.v) followed by the description of the output function in the 4 possible forms given by the Mealy-Moore and immediate-delayed distinctions.

Defines The problem solved by the finite automaton used as example is the detection of the sequence bb in the stream of symbols belonging to the set {a, b}.

The file describing the variables is the following:

```

/* *****
File name:      defines.vh
Circuit name:
Description:
***** */
// input codes
  'define a (1'b0)
  'define b (1'b1)
// internal state
  'define init_state (2'b00) // initial state
  'define one_b_state (2'b01) // one b received
  'define final_state (2'b10) // final state
// output codes
  'define no (1'b0) // no bb yet received
  'define yes (1'b1) // bb have been received

```

The binary codes associated to the input and output set are defined by the used of the design, while the binary codes associated to the internal states of the automaton are defined by the designer. It is very important that the designer has the freedom to associate the binary code according to its criteria. Designer criteria take into account, as we will see in the ?? section, optimization or even physical realizability criteria.

Half-Automaton Solving the problem of detecting the sequence bb is done at the level of the semiautomaton and is independent of the way the automaton reports the result on the outputs. For this reason, the semiautomaton can receive a separate description, which will be used in the final form of the design by inserting it into one of the 4 forms that the automaton can take depending on the user's requirements.

```

/* *****
File name:      halfAutomaton.v
Circuit name:   HA for double b detector
Description:    behavioral description of the half-automaton
                designed to detect 'bb' in a stream
                of symbols belonging to the set {a,b}
***** */
'include "defines.vh"
module halfAutomaton(  output reg [1:0] state ,
                      input      in      ,
                      input      reset   ,
                      input      clock );
// f: the state transition function
  always @(posedge clock)
    if (reset) state <= 'init_state;
    else
      case (state)
        'init_state : if (in == 'b) state <= 'one_b_state;
                      else          state <= 'init_state ;

```

```

        'one_b_state : if (in == 'b) state <= 'final_state ;
                      else          state <= 'init_state ;
        'final_state :          state <= 'final_state ;
        default      :          state <= 'init_state ;
    endcase
endmodule

```

For safety in operation, but also for an easy validation of the project, the automaton has an initial state, i.e., it is a strictly initial automaton in the `init_state` state.

Note the "friendly" way in which the description is made. The automaton's behavior can be read very easily due to the way in which we represented the behavior symbolically. We can make the following reading:

At reset = 1 the automaton goes into init_state. In init_state if b is received, then the automaton goes to the state in which a b was received: one_b_state, if not it remains in init_state. In one_b_state, if the same symbol b is received on the input, then the automaton goes to the final recognition state, final_state, if not then it returns to init_state to restart the search. In final_state it automatically remains blocked until a new signal reset reinitializes the search.

Immediate Moore To complete the project of the finite automaton, we must include the two previously defined files in the topmodule that provides the output signal. A first form is the one in which the automaton responds immediately strictly according to its internal state. The transition function of the output does not depend on the folded value of the input. It is the form of an immediate Moore type automaton whose description follows:

```

/* *****
File name:      immediateMooreAutomaton.v
Circuit name:   Double b detector
Description:    behavioral description of the Moore finite
                automaton designed to detect 'bb' in a stream
                of symbols belonging to the set {a,b}
***** */
#include "defines.vh"
module immediateMooreAutomaton( output reg out
                               input in
                               input reset, clock);

    wire [1:0] state ;

    halfAutomaton ha(state, in, reset, clock);

    // g: the output combinational transition function
    always @(state) case(state)
        'init_state : out = 'no ;
        'one_b_state : out = 'no ;
        'final_state : out = 'yes ;
    endcase
endmodule

```

```

                default      : out = 1'bx ;
            endcase
endmodule

```

In the Moore form, the automaton immediately responds with a latency of one clock cycle, signaling the appearance of the second b. This method of implementation is the simplest, having the disadvantage of an output signal that can be loaded by parasitic transitions due to the hazard phenomenon. Sometimes, the latency of a cycle can be a problem.

Delayed Moore The output signal will be able to be cleaned in a radical way from the phenomena of combinational hazard by opting for the delayed Moore version. The output register will work as a pipeline register and allow a "clean" signal synchronized with the system clock. The price paid for this advantage is the increase in latency by one more unit. For the delayed version there is the following code:

```

/* *****
File name:      delayedMooreAutomaton.v
Circuit name:   An example of Moore-type automaton
Description:    behavioral description of the delayed Moore
                finite automaton designed to detect 'bb' in
                streams of symbols belonging to the set {a,b}
***** */
`include "defines.vh"
module delayedMooreAutomaton(    output reg out      ,
                                input  in          ,
                                input  reset, clock);

    wire [1:0] state ;

    halfAutomaton ha(state , in , reset , clock);

    // g: the delayed transition function
    always @(posedge clock) case(state)
        'init_state : out <= 'no ;
        'one_b_state : out <= 'no ;
        'final_state : out <= 'yes;
        default      : out <= 1'bx;
    endcase
endmodule

```

The delayed Moore version is the simplest and the more robust implementation of the detector we design. the output circuit is simple, because it do not depend by input, and the output signal is easy to use because i fast and clean. It is recommended if the two-cycle latency can be "absorbed" in the system design.

Immediate Mealy If we are looking for the fastest response of the automaton, the Mealy immediate version is the solution. But the price we will pay is not negligible:

- the output signal depends on the temporal behavior of the circuit that generates the input of the automaton we are designing
- the combinational hazard cannot be eliminated
- the circuit that calculates the output function, g , is larger and more complex.

Whenever possible, this version should be avoided. We have an extra chance when we use it if we control the whole system in which the machine works.

```

/* *****
File name:      immediateMealyAutomaton.v
Circuit name:   An example of Mealy-type automaton
Description:    behavioral description of the Mealy finite
                automaton designed to detect 'bb' in a
                stream of symbols belonging to the set {a,b}
***** */
#include "defines.vh"
module immediateMealyAutomaton (output reg out
                               ,
                               input  in
                               ,
                               input  reset , clock);

    wire [1:0] state ;

    halfAutomaton ha(state , in , reset , clock);

// g: the output combinational transition function
    always @(*)
        case (state)
            'init_state : out = 'no ;
            'one_b_state : if (in == 'b) out = 'yes;
                        else out = 'no ;
            'final_state : out = 'yes;
            default     : out = 1'bx;
        endcase
endmodule

```

The set-up time of the input signal is defined only for the state transition function, f , because for the output transition function it is not possible because of the combinational nature of the function g .

Delayed Mealy The situation starts to become more controllable in the case of the delayed Mealy version. There still remains the problem of the set-up time, which must be defined both with respect to the state register and with respect to the output pipeline register.

```

/* *****
File name:      mealy_delayed_automaton.v
Circuit name:   An example of Mealy-type automaton
Description:    behavioral description of the Mealy finite

```

```

        automaton designed to detect 'bb' in a
        stream of symbols belonging to the set {a,b}
        *****/
'include "defines.vh"
module delayedMealyAutomaton(    output reg out        ,
                                input  in            ,
                                input  reset , clock );

    wire [1:0] state ;

    halfAutomaton ha(state , in , reset , clock);

// g: the delayed transition function
    always @(posedge clock)
        case(state)
            'init_state : out <= 'no ;
            'one_b_state : if (in == 'b) out <= 'yes ;
                        else out <= 'no ;
            'final_state : out <= 'yes ;
            default      : out <= 1'bx ;
        endcase
endmodule

```

From the point of view of latency, this version behaves the same as the immediate Moore automaton, but has the advantage of a synchronous output with the system clock.

4.2.2 Designing Finite Automata

Preliminary Examples

The behavior of a finite automaton can be defined in many ways. Graphs, transition tables, flow-charts, transition V/K diagrams or HDL description are very good for defining the transition functions f and g . All these forms provide non-recursive definitions. Thus, the resulting automata has the size of the definition in the same order with the size of the structure. Therefore, the finite automata are complex structures even when they have small size.

In order to exemplify the design procedure for a finite automaton let us take two examples, one dealing with a 1-bit input string and another related with a system built around the *multiply-accumulate circuit* (MAC) previously described.

Example 4.9 *The binary strings 1^n0^m , for $n \geq 1$ and $m \geq 1$, are recognized by a finite half-automaton by its internal states. Let's define and design it. The transition diagram defining the behavior of the half-automaton is presented in Figure 4.10, where:*

- q_0 - is the initial state in which 1 must be received, if not the half-automaton switches in q_3 , the error state
- q_1 - in this state at least one 1 was received and the first 0 will switch the machine in q_2
- q_2 - this state acknowledges a well formed string: one or more 1s and at least one 0 are already received

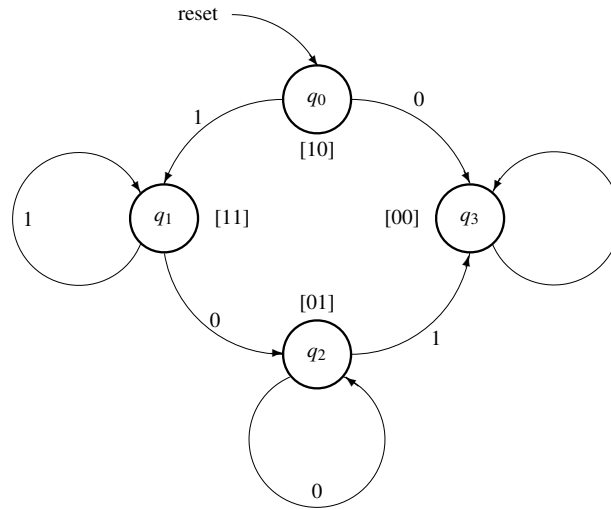


Figure 4.10: **Transition diagram.** The transition diagram for the half-automaton which recognizes strings of form $1^n 0^m$, for $n \geq 1$ and $m \geq 1$. Each circle represent a state, each (marked) arrow represent a (conditioned) transition.

- q_3 - the error state: an incorrect string was received.

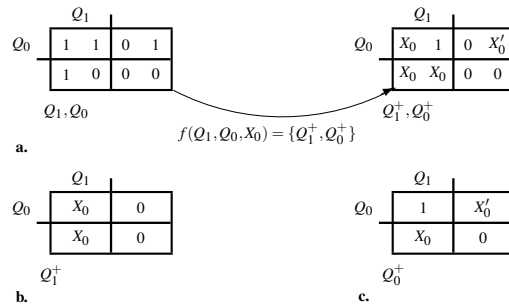


Figure 4.11: **VK transition maps.** The VK transition map for the half-automaton used to recognize $1^n 0^m$, for $n \geq 1$ and $m \geq 1$. **a.** The state transition function f . **b.** The VK diagram for the next most significant state bit, extracted from the previous full diagram. **c.** The VK diagram for the next least significant state bit.

The first step in implementing the structure of the just defined half-automaton is to **assign binary codes** to each state.

In this stage we have the absolute freedom. Any assignment can be used. The only difference will be in the resulting structure but not in the resulting behavior.

For a first version let be the codes assigned int square brackets in Figure 4.10. Results the transition diagram presented in Figure 4.11. The resulting transition functions are:

$$Q_1^+ = Q_1 \cdot X_0 = ((Q_1 \cdot X_0)')$$

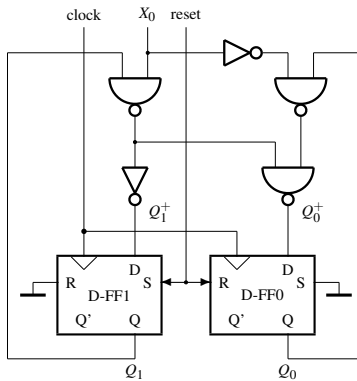


Figure 4.12: **A 4-state finite half-automaton.** The structure of the finite half-automaton used to recognize binary string belonging to the $1^n 0^m$ set of strings.

$$Q_0^+ = Q_1 \cdot X_0 + Q_0 \cdot X_0' = ((Q_1 \cdot X_0)' \cdot (Q_0 \cdot X_0'))'$$

(The 1 from q_0^+ map is double covered. Therefore, it is taken into consideration as a “don’t care”.) The circuit is represented in Figure 4.34 in a version using inverted gated only. The 2-bit state register is designed by 2 D flip-flops. The reset input is applied on the set input of D-FF1 and on the reset input of D-FF0.

The Verilog behavioral description of the automaton is:

```

/*****
File name:      rec_aut.v
Circuit name:   Recognizing Automaton for streams of form a^nb^m
Description:    behavioral description of the automaton used to recognize
streams of symbols of form a^nb^m
*****/
module rec_aut( output reg [1:0] state ,
input in ,
input reset ,
input clock );
always @(posedge clock)
if (reset) state <= 2'b10;
else case(state)
2'b00: state <= 2'b00 ;
2'b01: state <= {1'b0, ~in} ;
2'b10: state <= {in, in} ;
2'b11: state <= {in, 1'b1} ;
endcase
endmodule

```

◇

Example 4.10 Let us revisit the previous example in a more accurate implementation. Now a stream

of characters to be recognized is delimited by the empty character *e*. Therefore an actual stream to be recognized has the form:

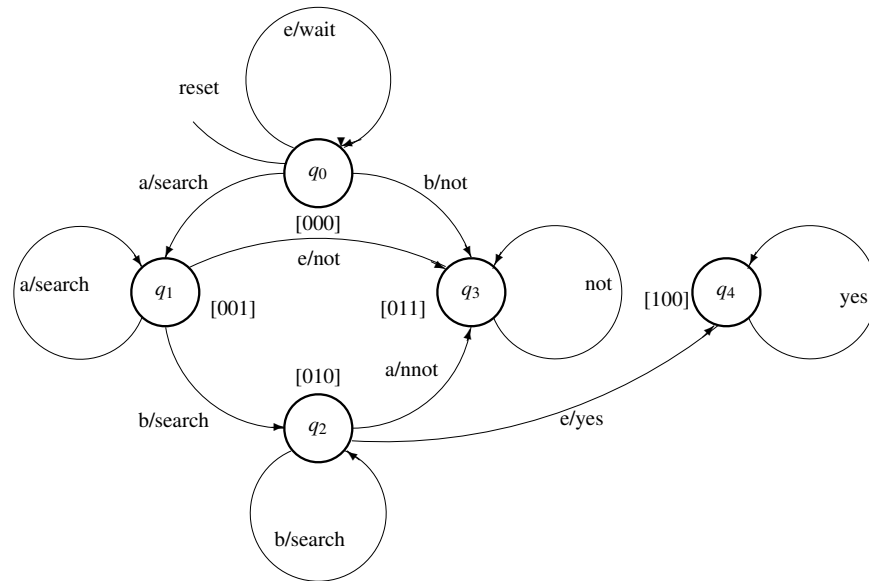


Figure 4.13:

... e e a a ... a b b ... b e e ...

The stream is considered recognized only when it ends. The graph describing the automaton has one state more compared with the previous approach, without the delimiting symbol *e*. It is represented in Figure 4.13. The automaton has the following 5 states:

- q_0 : the initial state in which the automaton goes by reset, and if
 - in = a** the automaton switches in q_1 signaling that it entered in the search state
 - in = b** the automaton switches in q_3 signaling that the stream started wrong and the search process failed
 - in = e** the automaton remains in q_0 waiting the start of an input stream of as and bs
- q_1 : the state waiting the flow of as
- q_2 : the state waiting the flow of bs
- q_3 : the state indicating that the string does not belong to the set $1^n 0^m | n, m \geq 1$
- q_4 : the state indicating that the string belongs to the set $1^n 0^m | n, m \geq 1$

The symbols used to describe the automaton are binary coded as follows:

- X = {a, b, e} = {01, 10, 00}
- Y = {wait, search, not, yes} = {00, 11, 01, 10}
- Q = {q0, q1, q2, q3, q4} = {000, 001, 010, 011, 100}

q2	q1	q0	x1	x0	q2+	q1+	q0+	y1	y0
0	0	0	0	0	0	0	0	0	0
0	0	0	0	1	0	0	1	1	1
0	0	0	1	0	0	1	1	0	1
0	0	0	1	1	-	-	-	-	-
0	0	1	0	0	0	1	1	0	1
0	0	1	0	1	0	0	1	1	1
0	0	1	1	0	0	1	0	1	1
0	0	1	1	1	-	-	-	-	-
0	1	0	0	0	1	0	0	1	0
0	1	0	0	1	0	1	1	0	1
0	1	0	1	0	0	1	0	1	1
0	1	0	1	1	-	-	-	-	-
0	1	1	0	0	0	1	1	0	1
0	1	1	0	1	0	1	1	0	1
0	1	1	1	0	0	1	1	0	1
0	1	1	1	1	-	-	-	-	-
1	0	0	0	0	1	0	0	1	0
1	0	0	0	1	1	0	0	1	0
1	0	0	1	0	1	0	0	1	0
1	0	0	1	1	-	-	-	-	-
1	0	1	0	0	-	-	-	-	-
⋮	⋮	⋮	⋮	⋮	-	-	-	-	-

Table 4.1: The truth table for the transition functions.

The sets X and Y are defined by the user (the one who proposed the design), while the state coding is at the discretion of the designer. Then, the Table 4.1 describing the state transition function and the output transition function.

We have to solve 5 functions of 5 variables. Let us use V-K diagrams for 4 variables (q_2 , q_1 , q_0 , x_1) and the 5th variable, x_0 , will be used to define the value of some boxes belonging to the diagrams. In Figure 4.14, we represented first the reference diagram to help us in defining the diagrams for f and g . We will explain at length how the diagram for the function q_2+ is built:

- in the box 0 is filled with 0, because for $\{q_2, q_1, q_0, x_1\} = \{0\ 0\ 0\ 0\}$ the output q_2+ does not depend on x_0 and takes the value 0
- in the box 1 is filled with 0, because for $\{q_2, q_1, q_0, x_1\} = \{0\ 0\ 0\ 1\}$ the output q_2+ could be considered 0 if we decide to select for the don't care value the value 0
- in the box 2 we fill up as in the box 0
- in the box 3 we fill up as in the box 1
- in the box 4 is filled with x_0 , because for $\{q_2, q_1, q_0, x_1\} = \{0\ 1\ 0\ 0\}$ the output q_2+ takes the value 1, if $x_0 = 0$ and the value 0 if $x_0 = 1$

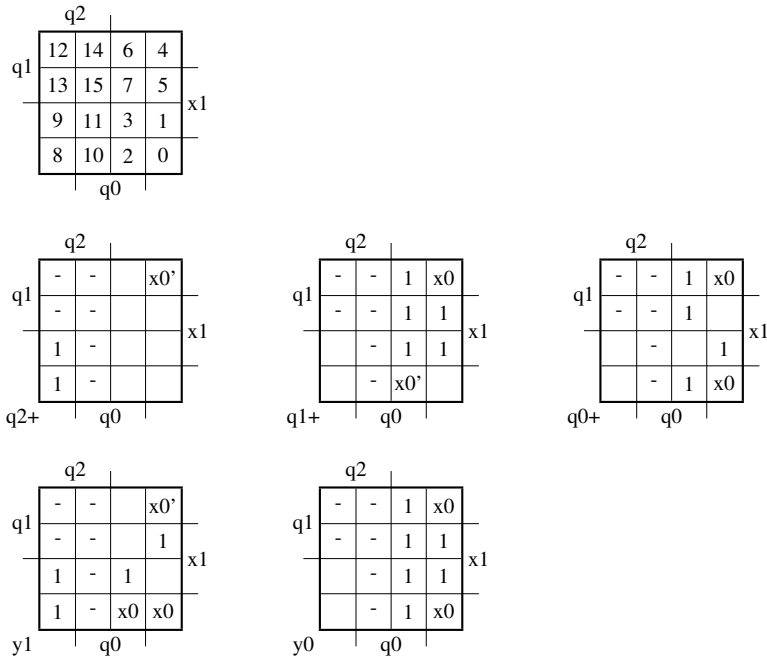


Figure 4.14: The V-K diagrams for the state and output transition functions.

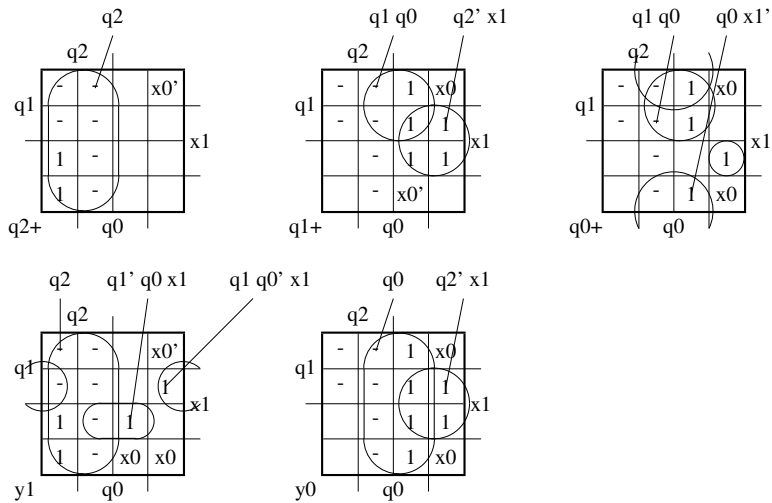


Figure 4.15: The first stage in the extracting algebraic expressions from V-K diagrams: the functions included in diagrams are ignored.

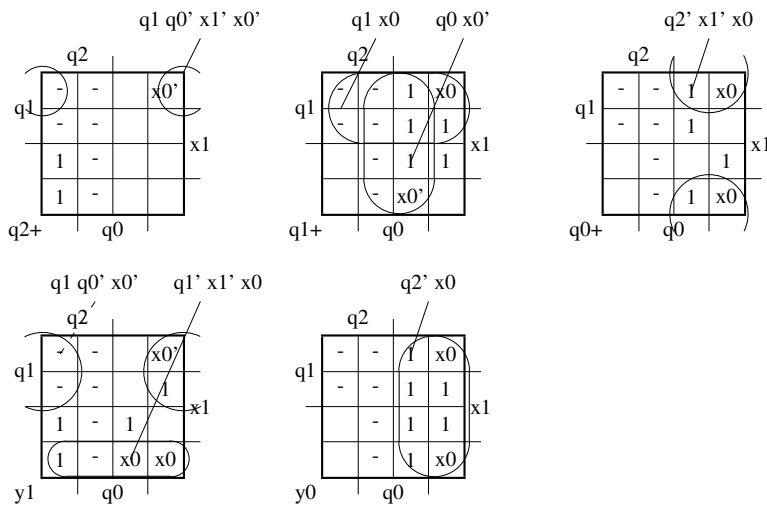


Figure 4.16: The second stage in the extracting algebraic expressions from V-K diagrams: the 1s are considered “don’t care”s.

- in the boxes 5 and 7 we do as for the box 1
- in the box 6 we do as for the box 0
- in the box 8 in the box 1, because for $\{q2, q1, q0, x1\} = \{1 0 0 0\}$ the output $q2+$ does not depend on $x0$ and takes the value 1
- in the box 9 in filled with 1, because for $\{q2, q1, q0, x1\} = \{1 0 0 1\}$ the output $q2+$ could be considered 1 if we decide to select for the don’t care value the value 1
- in the boxes 10 to 15 we fill up with don’t cares

The 5 function are extracted from the V-K diagrams in two stages. The first stage (which consider only the 1s from the diagram) is represented in Figure 4.15. The second stage (which considers the 1s as “don’t care”s) is represented in Figure 4.16 The resulting expressions are the following:

$$\begin{aligned}
 q2+ &= q2 + q1 q0' x1' x0' \\
 q1+ &= q2' x1 + q1 q0 + q0 x0' + q1 x0 \\
 q0+ &= q1 q0 + q0 x1' + q2' q1' q0' x1 + q2' x1' x0 \\
 y1 &= q2 + q1' q0 x1 + q1 q0' x1 + q1 q0' x0' + q1' x1' x0 \\
 y0 &= q0 + q2' x1 + q2' x0
 \end{aligned}$$

Until now we minimized each of the 5 functions independently. Each function is minimal, but what about the whole circuit? The global minimization supposes the maximization of the number of gates shared in the implementation of the 5 functions. Therefore, we must try to define the surfaces in the V-K diagram so as to maximize the number of identical surfaces, even if we will be pushed to avoid the minimal form for some functions.

In Figure 4.17 the diagram for $y0$ is modified: instead of the surface $q0$, emphasize in Figure 4.15, here we have a smaller one, $q0 x1'$, because this surface is selected also in the diagram for $q0+$. The impact on the final circuit is minimal: the fan-out of the D-FF0 is reduced.

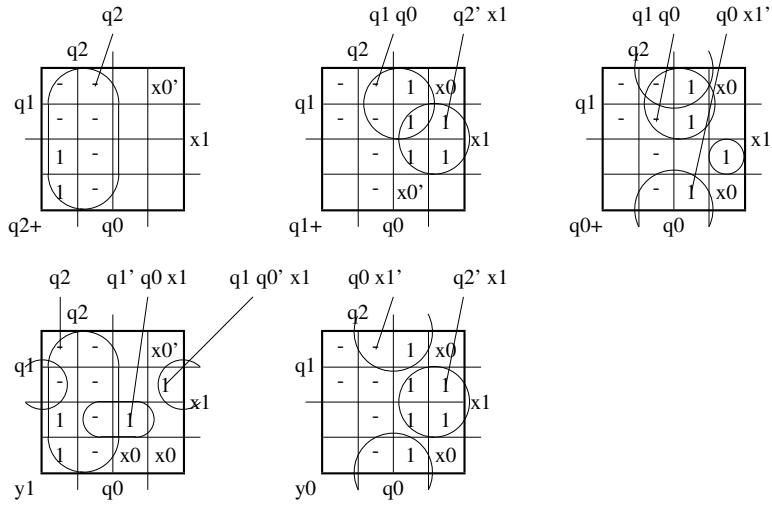


Figure 4.17: The first stage in the extracting algebraic expressions from V-K diagrams.

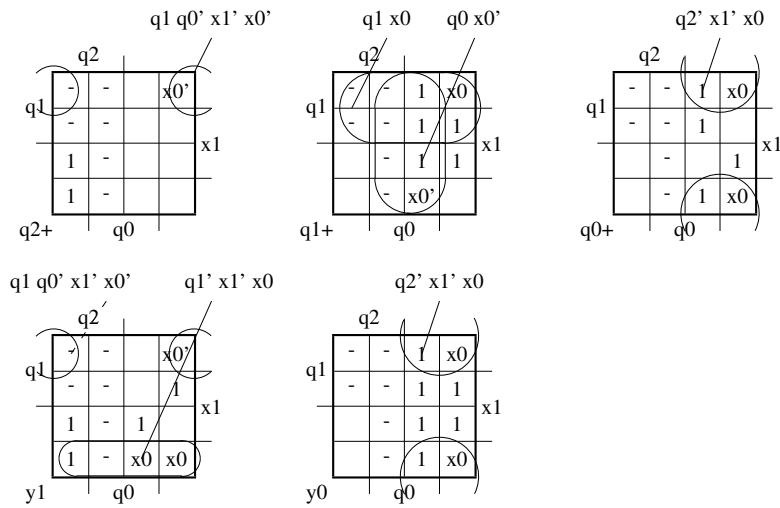


Figure 4.18: The second stage in the extracting algebraic expressions from V-K diagrams.

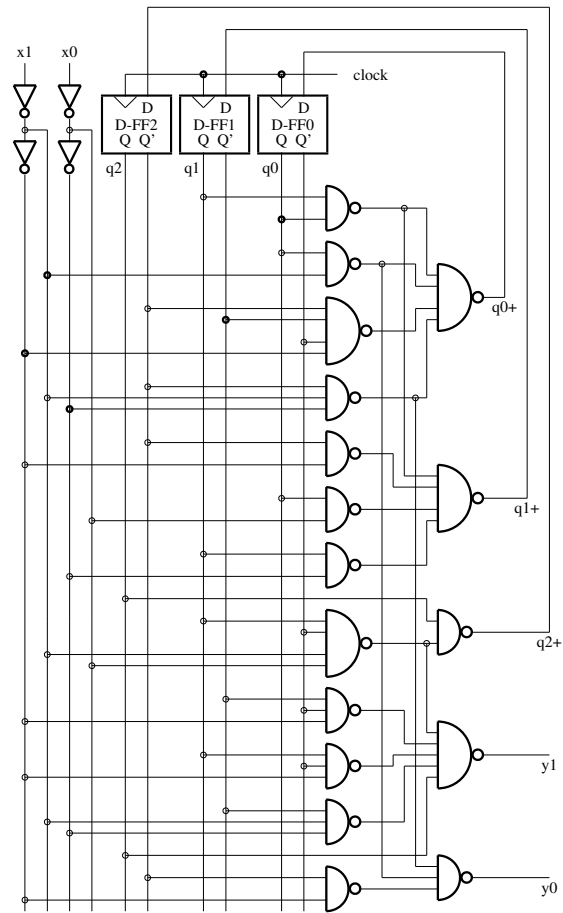


Figure 4.19: The resulting circuit.

The impact of this approach in the second stage is more important: the NAND circuit for $q_2' q_1' x_0$ is shared for the implementation of q_{0+} and y_0 , and the NAND circuit for $q_1 q_0' x_1' x_0'$ is shared for the implementation of q_{2+} and y_1 .

The resulting expressions are (with various brackets are emphasized the shared logic products):

$$\begin{aligned}
 q_{2+} &= q_2 + [q_1 q_0' x_1' x_0'] \\
 q_{1+} &= q_2' x_1 + \langle q_1 q_0 \rangle + q_0 x_0' + q_1 x_0 \\
 q_{0+} &= \langle q_1 q_0 \rangle + (q_0 x_1') + q_2' q_1' q_0' x_1 + \{q_2' x_1' x_0\} \\
 y_1 &= q_2 + q_1' q_0 x_1 + q_1 q_0' x_1 + [q_1 q_0' x_1' x_0'] + q_1' x_1' x_0 \\
 y_0 &= (q_0 x_1') + q_2' x_1 + \{q_2' x_1' x_0\}
 \end{aligned}$$

In Figure 4.19 is represented the resulting circuit, where the state register is implemented using 3 delay-flip-flops (D-FF) with their pair of outputs, one for Q and another for Q' . Thus, we do not need inverters for the bits coding the state. The circuit is implemented using NAND gates by applying the de Morgan law which transforms the AND-OR structure in a NAND-NAND configuration.

◇

Example 4.11 Let us revisit the previous example using another state coding:

$$Q = \{q_0, q_1, q_2, q_3, q_4\} = \{000, 001, 111, 011, 010\}$$

Then, the Table ?? describes the state transition function and the output transition function for the new coding.

The transition functions are represented with 3-variable V-K diagrams in Figure 4.20

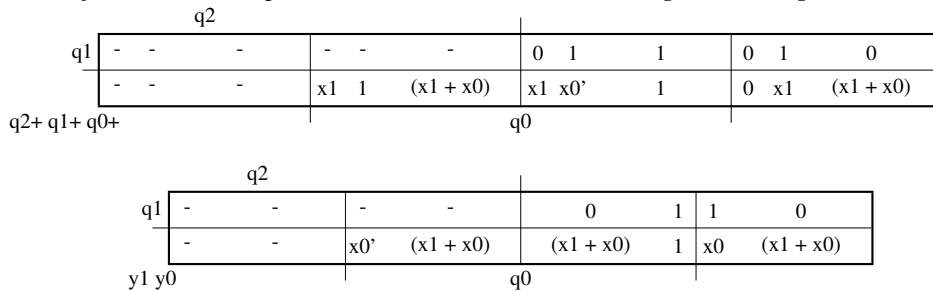


Figure 4.20:

From V-K diagrams result the following expressions :

$$\begin{aligned}
 q_{2+} &= q_1' q_0 x_1 \\
 q_{1+} &= q_2 + q_1 + q_0 x_0' + q_0' x_1 \\
 q_{0+} &= q_2' q_0 + q_1 (x_1 + x_0) \\
 y_1 &= q_1 q_0' + q_2 x_0' + q_0' x_0 + q_2' q_1' q_0 (x_1 + x_0) \\
 y_0 &= q_2' q_0 + q_1' (x_1 + x_0) = q_{0+}
 \end{aligned}$$

The resulting circuit is represented in Figure 4.21

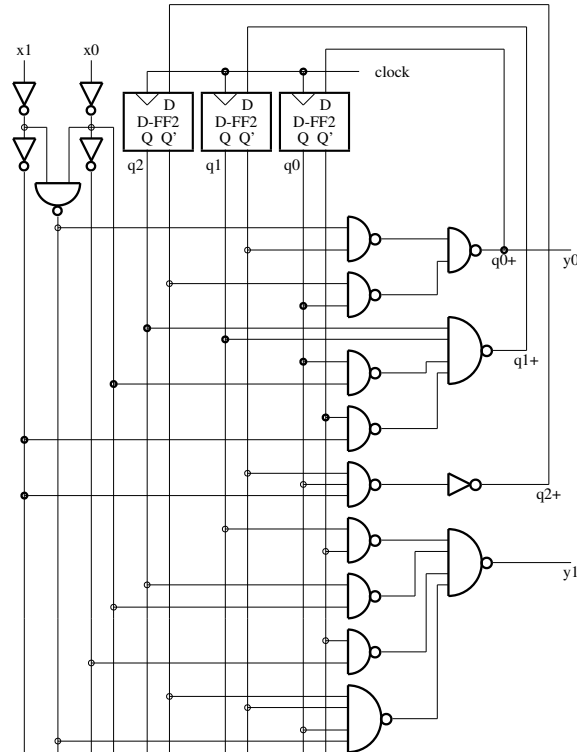


Figure 4.21: The circuit for the coding dominated by the reduce dependency coding style.

The size of the combinational circuits is only 70% from the previous solution. This reduction was obtained only by changing the state coding.

◇

The finite automaton has two distinct parts:

- the *simple, recursive defined part*, that consists in the state register; it can be minimized only by minimizing the definition of the automaton
- the *complex part*, that consists in the PLA that computes functions f and g and this is the part submitted to the main minimization process.

Our main goal in designing finite automaton is to reduce the random part of the automaton, even if the price is to enlarge the recursive defined part. In the current VLSI technologies *we prefer big size instead of big complexity*. A big sized circuit has now a technological solution, but for describing very complex circuits we have not yet efficient solutions (maybe never).

State Coding

The function performed by an automaton does not depend by the way its states are encoded, because the value of the state is a “hidden variable”. But, the actual structure of a finite automaton and its proper functioning are very sensitive to the state encoding.

The designer uses the freedom to code in different way the internal state of a finite automaton for its own purposes. A finite automaton is a concept embodied in physical structures. The transition from concept to an actual structure is a process with many traps and corner cases. Many of them are avoided using an appropriate coding style.

Example 4.12 Let be a first example showing the structural dependency by the state encoding. The automaton described in Figure 4.22a has three state. The first coding version for this automaton is: $q_0 = 00$, $q_1 = 01$, $q_2 = 10$. We compute the next state Q_1^+ , Q_0^+ , and the output Y_1 , Y_0 using the first two VK transition diagrams from Figure 4.22b:

$$\begin{aligned}
 Q_1^+ &= Q_0 + X_0 Q_1' \\
 Q_0^+ &= Q_1' Q_0' X_0' \\
 Y_1 &= Q_0 + X_0 Q_1' \\
 Y_0 &= Q_1' Q_0'.
 \end{aligned}$$

The second coding version for the same automaton is: $q_0 = 00$, $q_1 = 01$, $q_2 = 11$. Only the code for q_2 is different. Results, using the last two VK transition diagrams from Figure 4.22b:

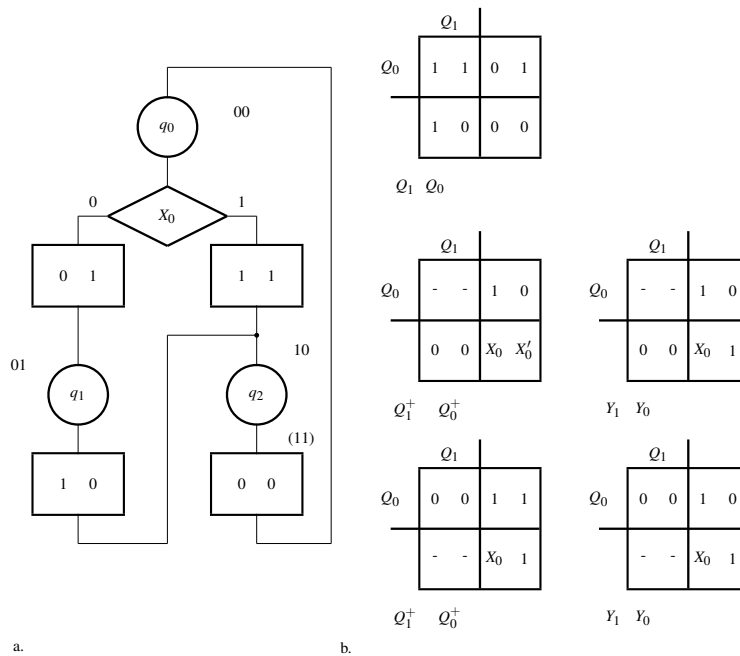


Figure 4.22: A 3-state automaton with two different state encoding. **a.** The flow-chart describing the behavior. **b.** The VK diagrams used to implement the automaton: the reference diagram for states, two transition diagrams used for the first code assignment, and two for the second state assignment.

$$\begin{aligned}
 Q_1^+ &= Q_1' Q_0 + X_0 Q_1' = (Q_1 + (Q_0 + X_0)')' \\
 Q_0^+ &= Q_1'
 \end{aligned}$$

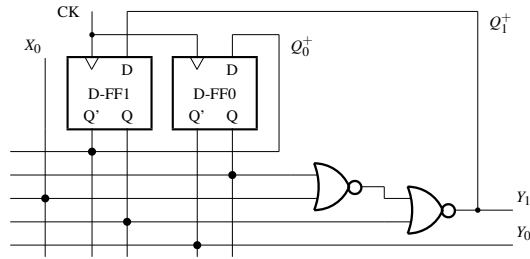


Figure 4.23: **The resulting circuit** It is done for the second state assignment of the automaton defined in Figure 4.22a.

$$Y_1 = Q_1'Q_0 + X_0Q_1' = (Q_1 + (Q_0 + X_0))'$$

$$Y_0 = Q_0'$$

Obviously the second coding version provides a simpler and smaller combinational circuit associated to the same external behavior. In Figure 4.23 the resulting circuit is represented. \diamond

Minimal variation encoding Minimal variation state assignment (or encoding) refers to the codes assigned to successive states.

Definition 4.8 *Coding with minimal variation means successive state are coded with minimal Hamming distance.* \diamond

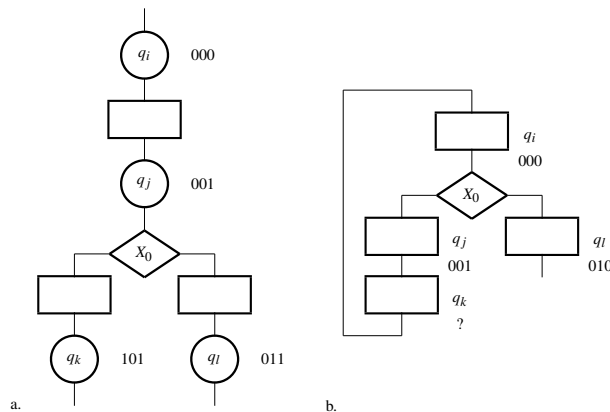


Figure 4.24: **Minimal variation encoding.** **a.** An example. **b.** An example where the minimal variation encoding is not possible.

Example 4.13 Let be the fragment of a flow chart represented in Figure 4.24a. The state q_i is followed by the state q_j and the assigned codes differ only by the least significant bit. The same for q_k and q_l which both follow the state q_j . \diamond

Example 4.14 *Some times the minimal variation encoding is not possible. An example is presented in Figure 4.24b, where q_k can not be coded with minimal variation.* \diamond

The minimal variation coding generates a minimal difference between the reference VK diagram and the state transition diagram. Therefore, the state transition logical function extracted from the VK diagram can be minimal.

Reduced dependency encoding Reduced dependency encoding refers to states which conditionally follow the same state. The reduced dependency is related to the condition tested.

Definition 4.9 *Reduced dependency encoding means the states which conditionally follow a certain state to be coded with binary configurations which differs minimal (have the Hamming distance minimal).* \diamond

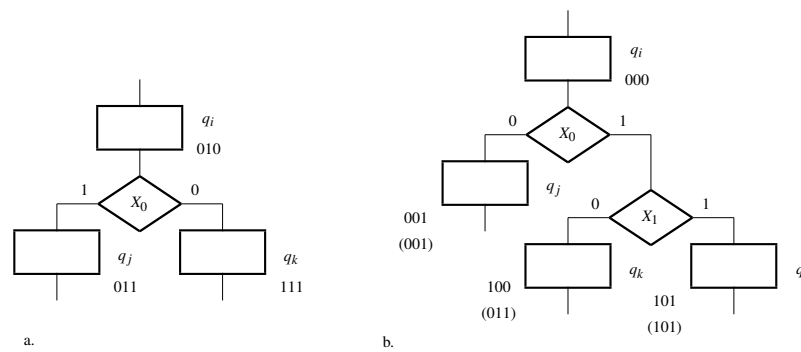


Figure 4.25: **Examples of reduced dependency encoding.** **a.** The transition from the state is conditioned by the value of a single 1-bit variable. **b.** The transition from the state is conditioned by two 1-bit variables.

Example 4.15 *In Figure 4.25a the states q_j and q_k follow, conditioned by the value of 1-bit variable X_0 , the state q_i . The assigned codes for the first two differ only in the most significant bit, and they are not related with the code of their predecessor. The most significant bit used to code the successors of q_i depends by X_0 , and it is X'_0 . We say: the next states of q_i are X'_011 , for $X_0=0$ the next state is 111, and for $X_0=1$ it is 011. Reduced dependency means: only one bit of the codes associated with the successors of q_i depends by X_0 , the variable tested in q_i .* \diamond

Example 4.16 *In Figure 4.25b the transition from the state q_i depends by two 1-bit variable, X_0 and X_1 . A reduced dependency coding is possible by only one of them. Without parenthesis is a reduced dependency coding by the variable X_1 . With parenthesis is a reduced dependency coding by X_0 .* \diamond

The reader is invited to provide the proof for the following theorem.

Theorem 4.2 *If the transition from a certain state depends by more than one 1-bit variable, the reduced dependency encoding can not be provided for more than one of them.* \diamond

The reduced dependency encoding is used to minimize the transition function because it allows to minimize the number of included variables in the VK state transition diagrams. Also, we will learn soon that this encoding style is very helpful in dealing with asynchronous input variables.

Incremental coding The incremental encoding provides an efficient encoding when we are able to use simple circuits to compute the value of the next state. An incremter is the simple circuit used to design the simple automaton called counter. The incremental encoding allows sometimes to center the implementation of a big half-automaton on a presetable counter.

Definition 4.10 *Incremental encoding means to assign, whenever it is possible, for a state following q_i a code determined by incrementing the code of q_i .* \diamond

Incremental encoding can be useful for reducing the complexity of a big automaton, even if sometimes the price will be to increase the size. But, as we more frequently learn, bigger size is a good price for reducing complexity.

One-hot state encoding The register is the simple part of an automaton and the combinational circuits computing the state transition function and the output function represent the complex part of the automaton. More, the speed of the automaton is limited mainly by the size and depth of the associated combinational circuits. Therefore, in order to increase the simplicity and the speed of an automaton we can use a coding style which increase the dimension of the register reducing in the same time the size and the depth of the combinational circuits. Many times a good balance can be established using the *one-hot state encoding*.

Definition 4.11 *The one-hot state encoding associates to each state a bit, and consequently the state register has a number of flip-flops equal with the number of states.* \diamond

All previous state encodings used a log-number of bits to encode the state. The size of the state register will grow, using one-hot encoding, from $O(\log n)$ to $O(n)$ for an n -state finite automaton. Deserves to pay sometimes this price for various reasons, such as speed, signal accuracy, simplicity, ...

Minimizing finite automata

There are formal procedure to minimize an automaton by minimizing the number of internal states. All these procedures refer to the concept. When the conceptual aspects are solved remain the problems related with the minimal physical implementation. Follow a short discussion about minimizing the size and about minimizing the complexity.

Minimizing the size by an appropriate state coding There are some simple rules to be applied in order to generate the possibility to reach a minimal implementation. Applying all of these rules is not always possible or an easy task and the result is not always guarantee. But it is good to try to apply them as much as possible.

A secure and simple way to optimize the state assignment process is to evaluate all possible coding versions and to choose the one which provide a minimal implementation. But this is not an effective way to solve the problem because the number of different versions is in $O(n!)$. For this reason are very useful some simple rules able to provide a good solution instead of an optimal one.

A lucky, inspired, or trained designer will discover an almost optimal solution applying the following rule in the order they are enounced.

Rule 1 : apply the reduced dependency coding style whenever it is possible. This rule allows a minimal occurrence of the input variable in the VK state transition diagrams. Almost all the time this

minimal occurrence has as the main effect reducing the size of the state transition combinational circuits.

Rule 2 : the states having the same successor with identical test conditions (if it is the case) will have assigned adjacent codes (with the Hamming distance 1). It is useful because brings in adjacent locations of a VK diagrams identical codes, thus generating the conditions to maximize the arrays defined in the minimizing process.

Rule 3 : apply minimal variation for unconditioned transitions. This rule generates the conditions in which the VK transition diagram differs minimally from the reference diagram, thus increasing the chance to find bigger surfaces in the minimizing process.

Rule 4 : the states with identical outputs are coded with minimal Hamming distance (1 if possible). Generates similar effects as Rule 2.

To see at work these rules let's take an example.

Example 4.17 *Let be the finite automaton described by the flow-chart from Figure 4.26. Are proposed two coding versions, a good one (the first), using the coding rules previously listed, and a bad one (the second with the codes written in parenthesis), ignoring the rules.*

For the first coding version results the expressions:

$$Q_2^+ = Q_2 Q_0' + Q_2' Q_1$$

$$Q_1^+ = Q_1 Q_0' + Q_2' Q_1' Q_0 + Q_2' Q_0 X_0$$

$$Q_0^+ = Q_0' + Q_2' Q_1' X_0'$$

$$Y_2 = Q_2 + Q_1 Q_0$$

$$Y_1 = Q_2 Q_1 Q_0' + Q_2' Q_1'$$

$$Y_0 = Q_2 + Q_1' + Q_0'$$

the resulting circuit having the size $S_{CLC_{ver1}} = 37$.

For the second coding version results the expressions:

$$Q_2^+ = Q_2 Q_1 Q_0' + Q_1' Q_0 + Q_2' Q_0 X_0 + Q_1 Q_0' X_0'$$

$$Q_1^+ = Q_1' Q_0 + Q_2' Q_1' + Q_2' X_0'$$

$$Q_0^+ = Q_1' Q_0 + Q_2' Q_1' + Q_2' X_0$$

$$Y_2 = Q_2 Q_0' + Q_2 Q_1 + Q_2' Q_1' Q_0 + Q_1 Q_0'$$

$$Y_1 = Q_2' Q_0 + Q_2' Q_1'$$

$$Y_0 = Q_2 + Q_1' + Q_0$$

the resulting circuit having the size $S_{CLC_{ver2}} = 50$. \diamond

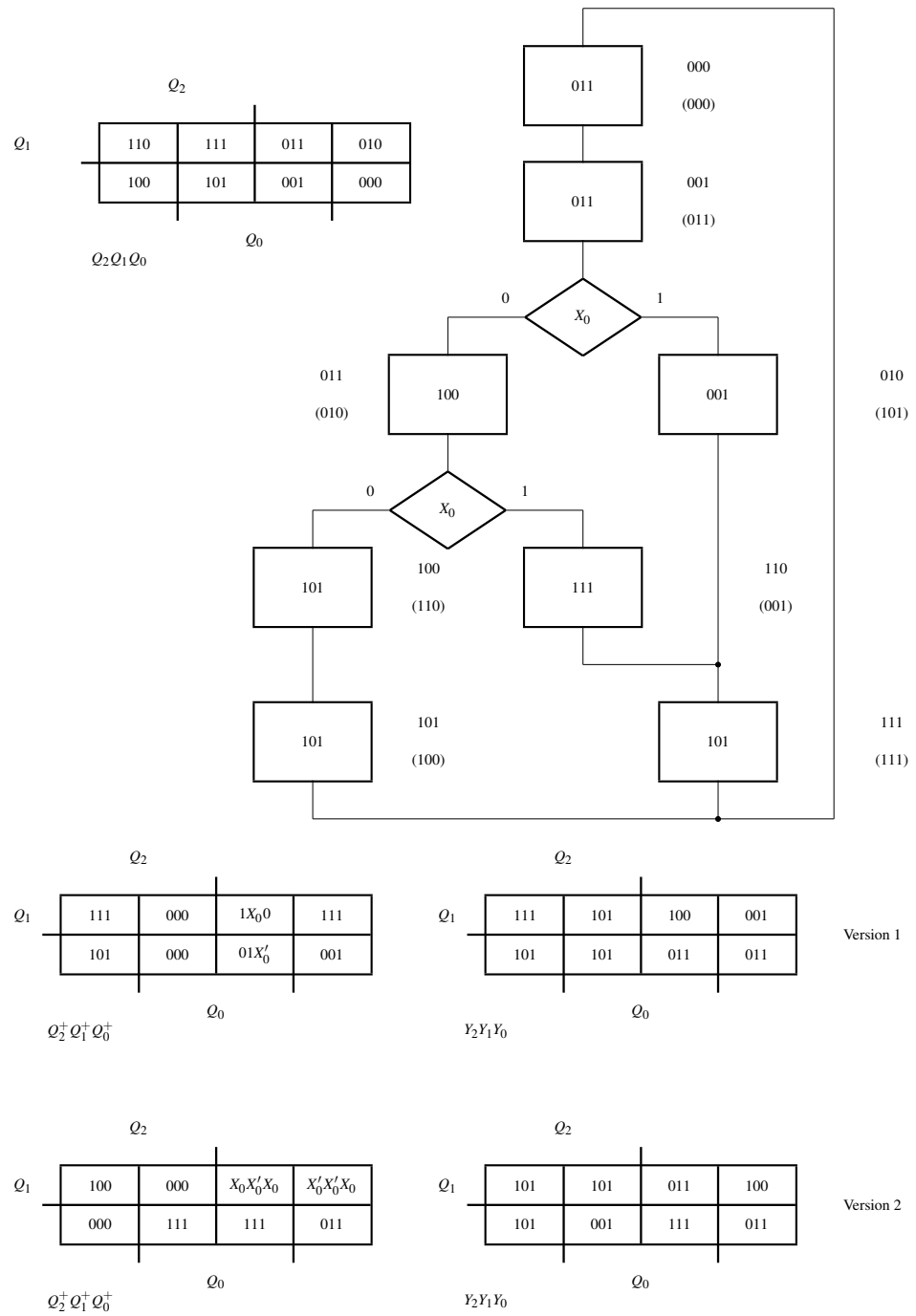


Figure 4.26: **Minimizing the structure of a finite automaton.** Applying appropriate coding rules the occurrence of the input variable X_0 in the transition diagrams can be minimized, thus resulting smaller Boolean forms.

Minimizing the complexity by one-hot encoding Implementing an automaton with one-hot encoded states means increasing the simple part of the structure, the state register. It is expected at least a part of this additional structure to be compensated by a reduced combinational circuit used to compute the transition functions. But, for sure the entire complexity is reduced because of a simpler combinational circuit.

Example 4.18 Let be the automaton described by the flow-chart from Figure 4.27, for which two coding version are proposed: a one-hot encoding using 6 bits ($Q_6 \dots Q_1$), and a compact binary encoding using only 3 bits ($Q_2 Q_1 Q_0$).

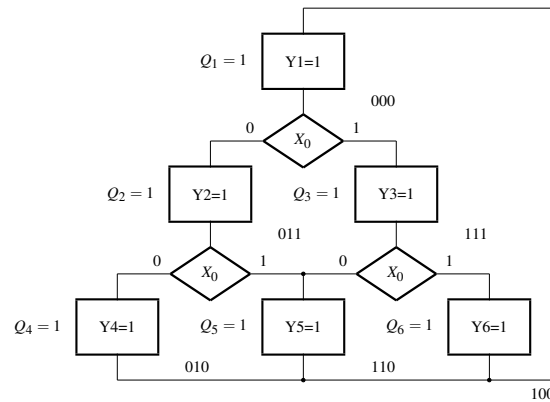


Figure 4.27: Minimizing the complexity using one-hot encoding.

The outputs are Y_6, \dots, Y_1 each active in a distinct state.

Version 1: with "one-hot" encoding The state transition functions, Q_i^+ , $i = 1, \dots, Q_6^+$, can be written directly inspecting the definition. Results:

$$Q^+_1 = Q_4 + Q_5 + Q_6$$

$$Q^+_2 = Q_1 X'_0$$

$$Q^+_3 = Q_1 X_0$$

$$Q^+_4 = Q_2 X'_0$$

$$Q^+_5 = Q_2 X_0 + Q_3 X'_0$$

$$Q^+_6 = Q_3 X_0$$

Because in each state only one output bit is active, results:

$$Y_i = Q_i, \text{ pentru } i = 1, \dots, 6.$$

The combinational circuit associated with the state transition function is very simple, and for outputs no circuits are needed. The size of the entire combinational circuit is $S_{CLC, \text{var}1} = 18$, with the big advantage that the outputs come directly from a flip-flop without additional unbalanced delays or other parasitic effects (like different kinds of hazards).

Version 2: compact binary coding The state transition functions for this coding version (see Figure 4.27 for the actual binary codes) are:

$$Q^+_2 = Q_2Q_0 + Q_0X_0 + Q'_2Q'_1X_0$$

$$Q^+_1 = Q'_2Q_0 + Q'_2Q'_1 + Q_0X'_0$$

$$Q^+_0 = Q'_2Q'_1$$

For the output transition function an additional decoder, DCD_3 , is needed. The resulting combinational circuit has the size $S_{CLC,var2} = 44$, with the additional disadvantage of generating the outputs signal using a combinational circuit, the decoder. \diamond

4.2.3 Control Automata (CROM)

When we are faced with problems that require a complex automaton of large dimensions, there is the possibility of segregating in its structure simple substructures that allow the total compactness of the system to be reduced. This is the case of some *automatic control devices* used, for example, as subsystems in microprogrammed systems.

A control automaton is included in a system using three main connections (see Figure 4.28):

- the p -bit input operation $[p-1:0]$ selects the control sequence to be executed by the control automaton (it *receives* the information about “what to do”); it is used to part the ROM in 2^p parts, each having the **same** dimension; in each part a sequence of maximum 2^n operation can be “stored” for execution
- the m -bit command output, $command [m-1:0]$, the control automaton uses to *generate* “the command” toward the controlled subsystem
- the n -bit input $flags [q-1:0]$ the control automaton uses to *receive* information, represented by some *independent bits*, about “what happens” in the controlled subsystems commanded by the output $command [m-1:0]$.

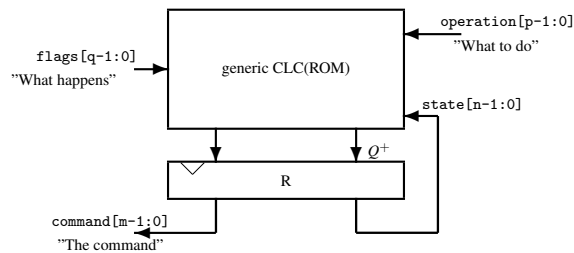


Figure 4.28: **Control Automaton.** The functional definition of control automaton. Control means to issue commands and to receive back signals (flags) characterizing the effect of the command.

Since, as we know, the maximum theoretical size of a random (complex) combinational circuit depends exponentially on the number of inputs, we must treat the number of inputs of the system in the generic version of Figure 6.3 very carefully. Some very useful observations can be made that allow the drastic reduction of the complex combinational circuit:

- the input operation [p-1:0] is considered only when a sequence of micro-instructions is initiated
- the inputs flags [q-1:0] have independent meaning and are considered independently at different times of the generation of microinstruction sequences
- an important share of the transitions of this automaton generates linear sequences of microinstructions, so the next state can be coded by incrementing the current one

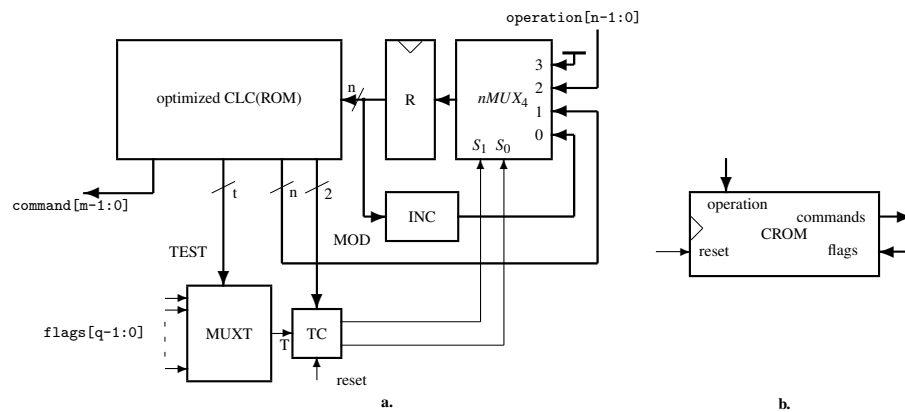


Figure 4.29: **The simplest Controller with ROM (CROM).** **a.** The Moore form of control automaton is optimized using an incremented circuit (INC) to compute the most frequent next address for ROM. **b.** The logic symbol for CROM.

In Figure 4.29 is represented an optimized form of the control automaton in which a series of simple circuits have been introduced that allow the minimization of the large and complex circuit represented by the combinational logic circuit, CLC, which can be implemented in the form of a ROM (Read - Only Memory). The role of these circuits is as follows:

MUXT : is used to select in each state only one flag from the set of q , because the current use of such a system thought us that the sequence of microinstructions depends, in most of cases, only by one flag in any cycle

$nMUX_4$: selects the transition mode of the automaton:

$S_1S_0 = 00$: the system is initialised in state $00\dots 0$

$S_1S_0 = 01$: the system takes one of its initial state corresponding to the microprogram

$S_1S_0 = 10$: the next state of the system is selected from the output of the random CLC

$S_1S_0 = 11$: the next state of the system is obtained by incrementing the value of the current state

INC : is the simple circuit of an increment circuit used to determine the next state of the system for the linear sequence of the microprogram

TC : is a very simple and small combinational circuit used to select the transition mode of the automaton according to the current state and of the flag selected by MUXT.

The size of optimized CLC(ROM) is dramatically reduced compared to the size of generic CLC(ROM) because, as we know, each remove of an input of a CLC reduces its theoretical size to half.

CROM is a very good example for using, whenever possible, the segregation of simplicity from an apparently very complex reality. With this circuit, we approach the class of automata circuits whose loop is mainly closed by simple, functional circuits.

4.3 Functional Automata: the Simple Automata

The smallest automata before presented are used in recursively extended configuration to perform similar functions for any n . From this category of circuits we will present in this section only the *binary counters*. The next circuit will be also a simple one, having the definition independent by size. It is a *sum-prefix automaton*. The last subject will be a multiply-accumulate circuit built with two simple automata serially connected.

4.3.1 The Smallest Automaton: the T Flip-Flop

The size and the complexity of an automaton depends at least on the dimension of the sets defining it. Thus, the smallest (and also the simplest) automaton has *two states*, $Q = \{0, 1\}$ (represented with one bit), *one-bit input*, $T = \{0, 1\}$, and $Q = Y$. The associated structure is represented in Figure 4.30, where is represented a circuit with one-bit input, T, having a one-bit register, a D flip-flop, for storing the 1-bit coded state, and a combinational logic circuit, CLC, for computing the function f .

What can be the meaning of an one-bit “message”, received on the input T, by a machine having only two states? We can “express” with the two values of T only the following things:

no op : $T = 0$ - the state of the automaton *remains the same*

switch : $T = 1$ - the state of the automaton *switches*.

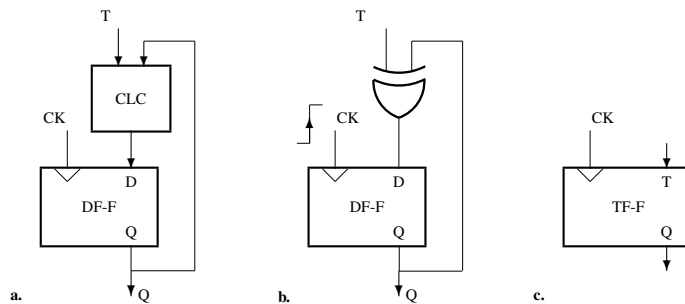


Figure 4.30: **The T flip-flop.** **a.** It is the simplest automaton because: has 1-bit state register (a DF-F), a 2-input loop circuit (one as automaton input and another to close the loop), and direct output from the state register. **b.** The structure of the T flip-flop: the XOR_2 circuits complements the state is $T = 1$. **c.** The logic symbol.

The resulting automaton is the well known *T flip-flop*. The actual structure of a T flip-flop is obtained connecting on the loop a commanded inverter, i.e., a XOR gate (see Figure 4.30b). The command input is T and the value to be inverted is Q, the state and the output of the circuit.

This small and simple circuit can be seen as a 2-modulo counter because for $T = 1$ the output “says”: 01010101... Another interpretation of this circuit is: the T flip-flop is a frequency divider. Indeed, if the clock frequency is f_{CK} , then the frequency of the signal received to the output Q is $f_{CK}/2$ (after each clock cycle the circuit comes back in the same state).

4.3.2 Counters

The first simple automaton is a composition starting from one of the function of T flip-flop: the counting. If one T flip-flop counts modulo-2¹, maybe two T flip-flops will count modulo-2² and so on. Seems to be right, but we must find the way for connecting many T flip-flops to perform the counter function.

For the synchronous counter² built with n T flip-flops, T_{n-1}, \dots, T_0 , the formal rule is very simple: if INC_0 , then the first flip-flop, T_0 , switches, and the i -th flip-flop, for $i = 1, \dots, n-1$, switches only if all the previous flip-flops are in the state 1. Therefore, in order to detect the switch condition for i -th flip-flop an AND_{i+1} must be used.

Definition 4.12 The n -bit synchronous counter, $COUNT_n$, has a clock input, CK , a command input, INC_0 , an n -bit data output, Q_{n-1}, \dots, Q_0 , and an expansion output, INC_n . If $INC_0 = 1$, the active edge of clock increments the value on the data output (see Figure 4.31). \diamond

There is also a recursive, constructive, definition for $COUNT_n$.

Definition 4.13 An n -bit synchronous counter, $COUNT_n$ is made by expanding a $COUNT_{n-1}$ with a T flip-flop with the output Q_{n-1} , and an AND_{n+1} , with the inputs $INC_0, Q_{n-1}, \dots, Q_0$, which computes INC_n (see Figure 4.31). $COUNT_1$ is a T flip-flop and an AND_2 with the inputs Q_0 and INC_0 which generates INC_1 . \diamond

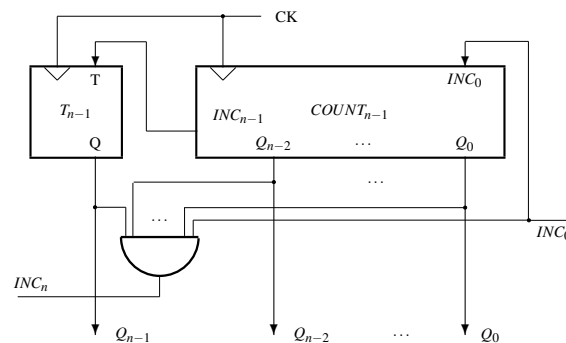


Figure 4.31: **The synchronous counter.** The recursive definition of a synchronous counter has $S_{COUNT}(n) \in O(n^2)$ and $T_{COUNT}(n) \in O(\log n)$, because for the i -th range one TF-F and one AND_i are added.

Example 4.19 *The Verilog description of a synchronous counter follows:

²There exist also asynchronous counters. They are simpler but less performant.

```

/* *****
File name:      sync_counter.v
Circuit name:   Synchronous Counter
Description:     structural description of a synchronous counter as a
                 T-type register loop connected with an AND prefix network
***** */
module sync_counter #(parameter n = 8)(output  [n-1:0] out      ,
                                     output      inc_n      ,
                                     input       inc_0      ,
                                     reset        ,
                                     clock       );

    t_reg    t_reg( .out      (out)          ,
                   .in       (prefix_out[n-1:0]) ,
                   .reset    (reset)         ,
                   .clock    (clock)         );

    and_prefix and_prefix( .out      (prefix_out)      ,
                           .in       ({out , inc_0})    );

    assign inc_n = prefix_out[n];
endmodule

```

```

/* *****
File name:      t_reg.v
Circuit name:   T-type Register
Description:     behavioral description of a register built using T-type
                 flip-flops instead of D-type flip flops
***** */
module t_reg #(parameter n = 8)( output  reg [n-1:0] out      ,
                                  input    [n-1:0] in        ,
                                  input    reset        ,
                                  clock     );

    always @(posedge clock) if (reset) out <= 0;
                                else   out <= out ^ in;

endmodule

```

The reset input is added because it is used in real applications. Also, a reset input is good in simulation because makes the simulation possible allowing an initial value for the flip-flops (`reg[n-1:0] out` in module `t_reg`) used in design. \diamond

It is obvious that $C_{COUNT}(n) \in O(1)$ because the definition for any n has the same, constant size (in number of symbols used to write the Verilog description for it or in the area occupied by the drawing of $COUNT_n$). The size of $COUNT_n$, according to the *Definition 4.4*, can be computed starting from the following iterative form:

$$S_{COUNT}(n) = S_{COUNT}(n-1) + (n+1) + S_T$$

and results:

$$S_{COUNT}(n) \in O(n^2)$$

because of the AND gates network used to command the T flip-flop. The counting time is the clock period. The minimal clock period is limited by the propagation time inside the structure. It is computed as follows:

$$T_{COUNT}(n) = t_{pT} + t_{pAND_n} + t_{SU} \in O(\log n)$$

where: $t_{pT} \in O(1)$ is the propagation time through the T flip-flop, $t_{pAND_n} \in O(\log n)$ is the propagation time through the AND_n (in the fastest version it is implemented using a tree of AND_2 gates) gate and $t_{SU} \in O(1)$ is the set-up time at the input of T flip-flop.

In order to reduce the size of the counter we must find another way to solve the function performed by the network of ANDs. Obviously, the network of ANDs is an *AND prefix-network*. Thus, the problem could be reduced to the problem of the general form of prefix-network. The optimal solution exists and has the size in $O(n)$ and the time in $O(\log n)$ (see in this respect the section 8.2).

Finishing this short discussion about counters must be emphasized the autonomy of this circuit which consists in switching in the next state *according to the current state*. We “tell” simply to the circuit “please count”, and the circuit know what to do. The loop allow “him to know” how to behave.

Real applications uses more complex counters able to be initialized in any states or the count in both ways, up and down. Such a counter is described by the following code:

```

/* *****
File name:      full_counter.v
Circuit name:   Full Counter
Description:    behavioral description of a counter with all the possible
                features (reset, load, up-count, down-count)
***** */
module full_counter #(parameter n = 4)(output reg [n-1:0] out ,
                                     input [n-1:0] in ,
                                     input reset ,
                                     load ,
                                     down ,
                                     count ,
                                     clock );

    always @(posedge clock)
        if (reset) out <= 0 ;
        else if (load) out <= in ;
        else if (count) if (down) out <= out - 1 ;
                        else out <= out + 1 ;
        else out <= out ;

endmodule

```

The reset operation has the highest priority, and the counting operations have the lowest priority.

Program Counter (PC)

Program Counter (PC) is a special counter can be used as a logic block in the structure of a processor. It control the evolution of the program execution. A version of a simple PC is represented in Figure 4.32, where:

- a 4-input multiplexor selects to the input of a register:
 - the incremented value of the PC stored in register in order to provide the address for the next instruction on the linear part of the program
 - the value of PC added with the jump address in order to allow program to perform a unconditional or a conditioned jump
 - the address of an absolute jump provided by the current instruction
 - the address for an absolute jump when a call or a return instruction is executed
- a zero detector combinational circuit provide the jump condition to
- a small random combinational circuit which generate the selection code for the multiplexor and the increment command for the increment circuit according the command `next` generated by one field of the instruction and according to the condition provided by zero and the interrupt signal `int`.

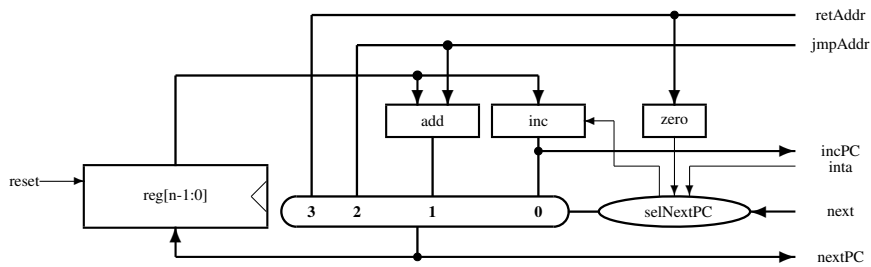


Figure 4.32: Program Counter

The transition set this circuit is controlled by the `selNextPC` combinational circuit described as follows::

```

/* *****
File: selNextPC.v
Name:
Description:
***** */
module selNextPC (
    output reg [1:0] sel ,
    output reg      inc ,
    input  [2:0] next ,
    input      zero ,
    input      int );

    always @(*)
        if (int)    {inc , sel} = 3'bx11 ;
        else

```

```

case ( next )
    // increment PC
    3'b000: {inc , sel} = 3'b100 ;
           // unconditional relative jump
    3'b001: {inc , sel} = 3'bx01 ;
           // unconditional absolute jump
    3'b010: {inc , sel} = 3'bx10 ;
           // return from subroutine
    3'b011: {inc , sel} = 3'bx11 ;
           // branch if zero
    3'b100: {inc , sel} = zero ? 3'bx01 : 3'b100 ;
           // branch if not zero
    3'b101: {inc , sel} = !zero ? 3'bx01 : 3'b100 ;
           // halt
    3'b110: {inc , sel} = 3'b000 ;
    default : {inc , sel} = 3'b100 ;
endcase
endmodule

```

In a real system, the connection of this circuit will be done taking into account the temporal relationships that must be optimized. As a rule, such relationships are established using pipeline registers.

4.3.3 Structured State Space Automaton(S^3A)

Definition 4.14 *The function:*

$$P(i, n, x_0, x_1, \dots, x_{n-1}) = x_i$$

is the projection (selection) function which returns the i -th element from a set of n elements.

◇

Definition 4.15 *A 3-port S^3A is defined by: $S^3A = (F \times X \times D \times L \times R; Y; \mathcal{S}; f, g)$ where:*

- $\mathcal{S} = (S_0 \times S_1 \times \dots, \times S_{m-1})$ with $S_i = \{0, 1\}^n$ for $i = 0, \dots, m-1$ is the structured state space
- $H = \{0, 1\}^{\log_2 P}$ is used to select a function from the set $\{h_0, h_1, \dots, h_p\}$
- $X = \{0, 1\}^n$ is the finite set of inputs binary represented on n bits
- $Y = (\{0, 1\}^n \times \{0, 1\}^n)$ is the finite set of outputs binary represented by two n -bit words
- $D = L = R = \{0, 1\}^{\log_2 m}$ are sets of pointers in the Cartesian product \mathcal{S}
- $g : (L \times R) \rightarrow (S_L \times S_R)$ is the output transition function
- $f : (H \times X \times D \times L \times R \times \mathcal{S}) \rightarrow S_D$ is the state transition function of form $h_H : (X, S_L, S_R) \rightarrow S_D$.

◇

An S^3A is implemented using a synchronous RAM to store the state. The inputs D, L, R are the address which select the elements of the Cartesian product stored in the m locations of the RAM. The efficiency of this approach could be evaluated as follows. The execution time for a full transition of S^3A is m times bigger than for the equivalent standard automaton, because only one element of the Cartesian product can be computed in one cycle. Therefore the time performance is $1/m$. The size of the combinational circuit for f belongs, in the worst case, to $O(2^{2n+\log_2 p})$, while for the standard automaton it belongs, in the worst case, to $O^{mn+\log_2 p}$. Results a decrease in size belonging to $O(2^{n(m-2)})$. The time performance decreases linearly with m , while the size decreases exponentially with m . There is no room for debate: when possible, the S^3A is the solution.

4.3.4 Multi-port S^3A

Because the binary functions dominate the class of arithmetic and logic functions, multi-port S^3A s are used in designing the executing core of any processing element. The most frequently used multi-port S^3A is a 3-port S^3A . Two ports are used to fetch the operands and the third for selecting the destination of the result. The following definition refers only the the half-automaton, because only the way the loop is closed is important. We can get the output of the system in various ways, depending on the application.

Definition 4.16 A 3-port Structured State Space Half-Automaton, S^3HA is defined as following:

$$S^3HA = (X \times DA \times LA \times RA, \mathbf{Q}, f)$$

where:

- $\mathbf{Q} = (Q_0 \times Q_1 \times \dots \times Q_{s-1})$: is the structured state space described as a Cartesian set of elements binary represented on m bits
- X : the finite set of inputs binary represented on p bits
- DA : the finite set of codes used to select the element of the set \mathbf{Q} to be modified (is the destination of the change) in the current state transition
- LA : the finite set of codes used to select the element of the set \mathbf{Q} to be used as left operand in the current state transition
- RA : the finite set of codes used to select the element of the set \mathbf{Q} to be used as right operand in the current state transition
- $f : (X \times LA \times RA \times \mathbf{Q}) = (X \times P(i, s, \mathbf{Q}) \times P(j, s, \mathbf{Q})) = (X \times Q_i \times Q_j) \rightarrow P(k, s, \mathbf{Q}) = Q_k$ is the state transition function where $i \in LA, j \in RA, k \in DA$

◇

Example 4.20 Let be a RALU designed with two modules already presented in the previous sections: the ALU exemplified in Example 2.4 and the register file presented in Simulation 3.13. In Figure 4.33 is represented the schematic of a 32-bit RALU.

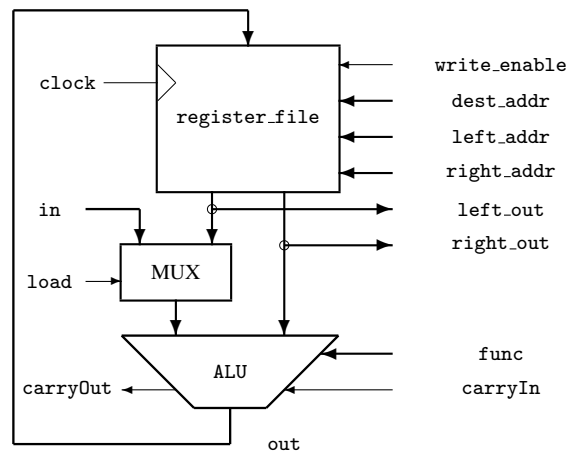


Figure 4.33: 32-bit RALU.

```

/* *****
File:          RALU.v
Circuit name:  RALU: Register file with Arithmetic and Logic Unit
Description:   register file with 16 32-bit register and an ALU with 8
              generic arithmetic and logic functions.
***** */
module RALU(
    output [31:0] left_out    ,
    output [31:0] right_out   ,
    output        carryOut   ,
    input         load        ,
    input [3:0]    left_addr   ,
    input [3:0]    right_addr  ,
    input [3:0]    dest_addr   ,
    input         write_enable ,
    input [31:0]  in           ,
    input         carryIn     ,
    input [2:0]   func         ,
    input         clock       );

    wire [31:0] out ;

    register_file rf(
        .left_operand    ( left_out    ),
        .right_operand   ( right_out   ),
        .result          ( out         ),
        .left_addr       ( left_addr   ),
        .right_addr      ( right_addr  ),
        .dest_addr       ( dest_addr   ),
        .write_enable    ( write_enable ),

```

```

        . clock          ( clock          ) );

    ALU alu ( . carryIn  ( carryIn        ),
             . func     ( func           ),
             . left     ( load ? in : left_out ),
             . right    ( right_out      ),
             . carryOut ( carryOut       ),
             . out      ( out            ) );

endmodule

```

◇

4.4 Concluding about automata

A new step is made in this chapter in order to increase the autonomous behavior of digital systems. The second loop looks justified by new useful behaviors.

Synchronous automata need non-transparent state registers The first loop, closed for gain the storing function, is applied carefully to obtain stable circuits. Tough restrictions can be applied (even number of inverting levels on the loop) because of the functional simplicity. The functional complexity of automata rejects any functional restrictions applied for the transfer function associated to loop circuits. The unstable behavior is avoided using non-transparent memories (registers) to store the state³. Thus, the state switches synchronized by clock. The output switches synchronously for delayed version of the implementation. The output is asynchronous for the immediate versions.

The second loop means the behavior's autonomy Using the first loop to store the state and the second to compute *any* transition function, a half-automaton is able to evolve in the state space. The evolution depends by state and by input. The state dependence allows an evolution even if the input is constant. Therefore, the automaton manifests its autonomy being able to behave, evolving in the state space, under constant input. An automaton can be used as “pure” generator of more or less complex sequence of binary configuration. the complexity of the sequence depends by the complexity of the state transition function. A simple function on the second loop determine a simple behavior (a *simple* increment circuit on the second loop transforms a register in a counter which generate the *simple* sequence of numbers in the strict increasing order).

Simple automata can have n states When we say n states, this means n can be very big, it is not limited by our ability to define the automaton, it is limited only by the possibility to implement it using the accessible technologies. A simple automata can have n states because the state register contains $\log n$ flip-flops, and its second loop contains a simple (constant defined) circuit having the size in $O(f(\log n))$. The simple automata can be big because they can be specified easy, and they can be generated automatically using the current software tools.

³Asynchronous automata are possible but their design is restricted by to complex additional criteria. Therefore, asynchronous design is avoided until stronger reason will force us to use it.

Complex automata have only finite number of states Finite number of states means: a number of states unrelated with the length (theoretically accepted as infinite) of the input sequence, i.e., the number of states is constant. The definition must describe the specific behavior of the automaton in each state. Therefore, the definition is complex having the size (at least) linearly related with the number of states. Complex automata must be small because they suppose combinational loops closed through complex circuits having the description in the same magnitude order with their size.

Control automata suggest the third loop Control automata evolve according to their state and they take into account the signals received from the controlled system. Because the controlled system receives commands from the same control automaton a third loop prefigures. Usually finite automata are used as control automata. Only the simple automata are involved directly in processing data.

An important final question: adding new loops the functional power of digital systems is expanded or only helpful features are added? And, if indeed new helpful features occur, who is helped by these additional features?

4.5 Problems

Problem 4.1 Draw the JK flip-flop structure (see Figure ??) at the gate level. Analyze the set-up time related to both edges of the clock.

Problem 4.2 Design a JK FF using a D flip-flop by closing the appropriate combinational loop. Compare the set-up time of this implementation with the set-up time of the version resulting in the previous problem.

Problem 4.3 Design the sequential version for the circuit which computes the n -bit AND prefixes. Follow the approach used to design the serial n -bit adder (see Figure ??).

Problem 4.4 Write the Verilog structural description for the universal 2-input, 2-state programmable automaton.

Problem 4.5 Draw at the gate level the universal 2-input, 2-state programmable automaton.

Problem 4.6 Use the universal 2-input, 2-state automaton to implement the following circuits:

- n -bit serial adder
- n -bit serial subtractor
- n -bit serial comparator for equality
- n -bit serial comparator for inequality
- n -bit serial parity generator (returns 1 if odd)

Problem 4.7 Define the synchronous n -bit counter as a simple n -bit Increment Automaton.

Problem 4.8 Design a Verilog tester for the resettable synchronous counter from Example 4.1.

Problem 4.9 Evaluate the size and the speed of the counter defined in Example 4.1.

Problem 4.10 Improve the speed of the counter designed in Example 4.1 designing an improved version for the module `and_prefix`.

Problem 4.11 Design a reversible counter defined as follows:

```

module smartest_counter    #(parameter n = 16)
    (
        output  [n-1:0] out    ,
        input   [n-1:0] in     , // preset value
        input   reset        , // reset counter to zero
        input   load         , // load counter with 'in'
        input   down         , // counts down if (count)
        input   count        , // counts up or down
        input   clock        );
    // ...
endmodule

```

Problem 4.12 Simulate a 3-bit counter with different delay on its outputs. It is the case in real world because the flop-flops can not be identical and their load could be different. Use it as input for a three input decoder implemented in two versions. One without delays and another assigning delays to the inverters and the the gates used to implement the decoder. Visualize the outputs of the decoder in both cases and interpret what you will find.

Solution:

```

/*****
File name:      dec_spyke.v
Circuit name:   Simulation module to emphasize the spyke to the output of
                decoder driven by a counter
Description:    describes a system with a clock generator, a counter and
                a decoder, in two versions: with delays and without
                delays associated to the gates
*****/
module dec_spyke;
    reg        clock ,
              enable ;
    reg [2:0]  counter ;
    wire       out0 , out1 , out2 , out3 , out4 , out5 , out6 , out7 ;

    initial begin    clock = 0;
                    enable = 1;
                    counter = 0;
                    forever #20 clock = ~clock;
                end

    initial #400 $stop;

    always @(posedge clock)
        begin
            counter[0] <= #3 ~counter[0];
            if (counter[0]) counter[1] <= #4 ~counter[1];
            if (&counter[1:0]) counter[2] <= #5 ~counter[2];
        end

    dmux dmux( .out0    (out0)    ,
               .out1    (out1)    ,
               .out2    (out2)    ,
               .out3    (out3)    ,
               .out4    (out4)    ,
               .out5    (out5)    ,
               .out6    (out6)    ,
               .out7    (out7)    ,
               .in      (counter)  ,
               .enable  (enable)  );

    initial $vw_dumpvars ;

endmodule

```

```

/* *****
File name:      dmux.v
Circuit name:   DMUX
Description:    structural description of a DMUX with and without delays
                associated to the gates
***** */
module dmux(out0 , out1 , out2 , out3 , out4 , out5 , out6 , out7 , in , enable);

    input      enable ;
    input  [2:0]  in ;
    output     out0 , out1 , out2 , out3 , out4 , out5 , out6 , out7 ;
// with no delay version
/*
    assign {out0 , out1 , out2 , out3 , out4 , out5 , out6 , out7} = 1'b1 << in ;
// */
// with delays version
/*
    not      #1  not0 (nin2 , in [2]);
    not      #1  not1 (nin1 , in [1]);
    not      #1  not2 (nin0 , in [0]);
    not      #1  not3 (in2 , nin2 );
    not      #1  not4 (in1 , nin1 );
    not      #1  not5 (in0 , nin0 );

    nand     #2  nand0 (out0 , nin2 , nin1 , nin0 , enable );
    nand     #2  nand1 (out1 , nin2 , nin1 , in0 , enable );
    nand     #2  nand2 (out2 , nin2 , in1 , nin0 , enable );
    nand     #2  nand3 (out3 , nin2 , in1 , in0 , enable );
    nand     #2  nand4 (out4 , in2 , nin1 , nin0 , enable );
    nand     #2  nand5 (out5 , in2 , nin1 , in0 , enable );
    nand     #2  nand6 (out6 , in2 , in1 , nin0 , enable );
    nand     #2  nand7 (out7 , in2 , in1 , in0 , enable );
// */
endmodule

```

Problem 4.13 Justify the reason for which the LIFO circuit works properly without a reset input, i.e., the initial state of the address counter does not matter.

Problem 4.14 How behaves `simple_stack`.

Problem 4.15 Design a LIFO memory using a synchronous RAM (SRAM) instead of an asynchronous one as in the embodiment represented in Figure ??.

Problem 4.16 Some applications ask the access to the last two data stored into the LIFO. Call them `tos`, for the last pushed data, and `prev_tos` for the previously pushed data. Both accessed data can be popped from stack. Double push is allowed. The accessed data can be rearranged swapping their position. Both, `tos` and `prev_tos` can be pushed again in the top of stack. Design such a LIFO defined as follows:

```

module two_head_lifo ( output [31:0] tos      ,
                      output [31:0] prev_tos  ,
                      input  [31:0] in       ,
                      input  [31:0] second_in ,
                      input  [2:0]  com      , // the operation
                      input          clock   );

// the semantics of 'com'
parameter  nop      = 3'b000, // no operation
            swap     = 3'b001, // swap the first two
            pop      = 3'b010, // pop tos
            pop2     = 3'b011, // pop tos and prev_tos
            push     = 3'b100, // push in as new tos
            push2    = 3'b101, // push 'in' and 'second_in'
            push_tos = 3'b110, // push 'tos' (double tos)
            push_prev = 3'b111; // push 'prev_tos'

// ...
endmodule

```

Problem 4.17 Write the Verilog description of the FIFO memory represented in Figure ??.

Problem 4.18 Redesign the FIFO memory represented in Figure ?? using a synchronous RAM (SRAM) instead of the asynchronous RAM.

Problem 4.19 There are application asking for a warning signal before the FIFO memory is full or empty. Sometimes full and empty come to late for the system using the FIFO memory. For example, no more then 3 write operation are allowed, or no more than 7 read operation are allowed are very useful in systems designed using pipeline techniques. The threshold for this warning signals is good to be programmable. Design a 256 8-bit entries FIFO with warnings activated using a programmable threshold. The interconnection of this design are:

```

module th_fifo ( output [7:0] out      ,
                 input  [7:0] in      ,
                 input  [3:0] write_th, // write threshold
                 input  [3:0] read_th , // read threshold
                 input          write ,
                 input          read  ,
                 output          w_warn, // write warning
                 output          r_warn, // read warning
                 output          full  ,
                 output          empty ,
                 input          reset  ,
                 input          clock  );

// ...
endmodule

```

Problem 4.20 A synchronous FIFO memory is written or read using the same clock signal. There are many applications which use a FIFO to interconnect two subsystems working with different clock signals. In this cases the FIFO memory has an additional role: to cross from the clock domain `clock_in` into another clock domain, `clock_out`. Design an **asynchronous FIFO** using a synchronous RAM.

Problem 4.21 A serial memory implements the data structure of a fix length circular list. The first location is accessed, for write or read operation, activating the input `init`. Each read or write operation move the access point one position right. Design an 8-bit word serial memory using a synchronous RAM as follows:

```

module serial_memory ( output [7:0] out      ,
                       input  [7:0] in       ,
                       input          init    ,
                       input          write   ,
                       input          read    ,
                       input          clock   );
endmodule

```

Problem 4.22 A list memory is a circuit in which a list can be constructed by `insert`, can be accessed by `read_forward`, `read_back`, and modified by `insert`, `delete`. Design such a circuit using two LIFOs.

Problem 4.23 Design a sequential multiplier using as combinational resources only an adder, a multiplexors.

Problem 4.24 Write the behavioral and the structural Verilog description for the MAC circuit represented in Figure ???. Test it using a special test module.

Problem 4.25 Redesign the MAC circuit represented in Figure ??? adding pipeline register(s) to improve the execution time. Evaluate the resulting speed performance using the parameters form Appendix E.

Problem 4.26 How many 2-bit code assignment for the half-automaton from Example 4.2 exist? Revisit the implementation of the half-automaton for four of them different from the one already used. Compare the resulting circuits and try to explain the differences.

Problem 4.27 Ad to the definition of the half-automaton from Example 4.2 the output circuits for: (1) `error`, a bit indicating the detection of an incorrectly formed string, (2) `ack`, another bit indicating the acknowledge of a well formed sting.

Problem 4.28 Multiplier control automaton can be defined testing more than one input variable in some states. The number of states will be reduced and the behavior of the entire system will change. Design this version of the multiply automaton and compare it with the circuit resulted in Example 4.3. Reevaluate also the execution time for the multiply operation.

Problem 4.29 Revisit the system described in Example 4.3 and design the finite automaton for multiply and accumulate (MACC) function. The system perform MACC until the input FIFO is empty and `end = 1`.

Problem 4.30 Design the structure of TC in the CROM defined in 4.4.3 (see Figure 4.29). Define the codes associated to the four modes of transition (jmp, cjmp, init, inc) so as to minimize the number of gates.

Problem 4.31 Design an easy to actualize Verilog description for the CROM unit represented in Figure 4.29.

Problem 4.32 Generate the binary code for the ROM described using the symbolic definition in Example 4.4.

Problem 4.33 Design a fast multiplier converting a sequential multiplier into a combinational circuit.

Problem 4.34 Let be the finite automaton defined in Figure 4.34. Do the following:

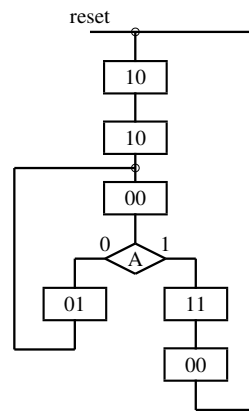


Figure 4.34:

1. assign the state codes in two versions:

- (a) according priority to the reduce dependency coding style
- (b) according priority to the minimal variation coding style

2. implement the finite automaton in the resulting two versions by:

- drawing the transition VK diagrams
- extracting the logic functions for Q_2^+ , Q_1^+ , Q_0^+ , Y_2 , Y_1 , Y_0
- drawing the logic schematic of the resulting automaton

Problem 4.35 Describe in Verilog the automaton defined in Problem 4.34 and simulate it.

4.6 Projects

Project 4.1 Finalize Project 1.2 using the knowledge acquired about the combinational and sequential structures in this chapter and in the previous two.

Project 4.2 The idea of simple FIFO presented in this chapter can be used to design an actual block having the following additional features:

- fully buffered inputs and outputs
- programmable thresholds for generating the empty and full signals
- asynchronous clock signals for input and for output (the design must take into consideration that the two clocks – `clockIn`, `clockOut` – are considered completely asynchronous)
- the read or write commands are executed only if it is possible (reads only if not-empty, or writes only if not-full).

The module header is the following:

```

module asyncFIFO #('include "fifoParameters.v")
  (
    output reg [n-1:0] out      ,
    output reg          empty   ,
    output reg          full    ,
    input      [n-1:0] in       ,
    input          write      ,
    input          read       ,
    input      [m-1:0] inTh    , // input threshold
    input      [m-1:0] outTh   , // output threshold
    input          reset      ,
    input          clockIn    ,
    input          clockOut );
  // ...
endmodule

```

The file `fifoParameters.v` has the content:

```

parameter    n = 16 , // word size
              m = 8  , // number of levels

```

Project 4.3 Design a stack execution unit with a 32-bit ALU. The stack is 16-level depth (`stack0`, `stack1`, ... `stack15`) with `stack0` assigned as the top of stack. ALU has the following functions:

- *add: addition*
 $\{\text{stack0}, \text{stack1}, \text{stack2}, \dots\} \leftarrow \{(\text{stack0} + \text{stack1}), \text{stack2}, \text{stack3}, \dots\}$
- *sub: subtract*
 $\{\text{stack0}, \text{stack1}, \text{stack2}, \dots\} \leftarrow \{(\text{stack0} - \text{stack1}), \text{stack2}, \text{stack3}, \dots\}$

- *inc: increment*
 $\{\text{stack0}, \text{stack1}, \text{stack2}, \dots\} \leq \{(\text{stack0} + 1), \text{stack1}, \text{stack2}, \dots\}$
- *dec: decrement*
 $\{\text{stack0}, \text{stack1}, \text{stack2}, \dots\} \leq \{(\text{stack0} - 1), \text{stack1}, \text{stack2}, \dots\},$
- *and: bitwise AND*
 $\{\text{stack0}, \text{stack1}, \text{stack2}, \dots\} \leq \{(\text{stack0} \& \text{stack1}), \text{stack2}, \text{stack3}, \dots\}$
- *or: bitwise OR*
 $\{\text{stack0}, \text{stack1}, \text{stack2}, \dots\} \leq \{(\text{stack0} | \text{stack1}), \text{stack2}, \text{stack3}, \dots\}$
- *xor: bitwise XOR*
 $\{\text{stack0}, \text{stack1}, \text{stack2}, \dots\} \leq \{(\text{stack0} \oplus \text{stack1}), \text{stack2}, \text{stack3}, \dots\}$
- *not: bitwise NOT*
 $\{\text{stack0}, \text{stack1}, \text{stack2}, \dots\} \leq \{(\sim \text{stack0}), \text{stack1}, \text{stack2}, \dots\}$
- *over:*
 $\{\text{stack0}, \text{stack1}, \text{stack2}, \dots\} \leq \{\text{stack1}, \text{stack0}, \text{stack1}, \text{stack2}, \dots\}$
- *dup: duplicate*
 $\{\text{stack0}, \text{stack1}, \text{stack2}, \dots\} \leq \{\text{stack0}, \text{stack0}, \text{stack1}, \text{stack2}, \dots\}$
- *rightShift: right shift one position (integer division)*
 $\{\text{stack0}, \text{stack1}, \dots\} \leq \{(\{1'b0, \text{stack0}[31:1]\}), \text{stack1}, \dots\}$
- *arithShift: arithmetic right shift one position*
 $\{\text{stack0}, \text{stack1}, \dots\} \leq \{(\{\text{stack0}[31], \text{stack0}[31:1]\}), \text{stack1}, \dots\}$
- *get: push dataIn in top of stack*
 $\{\text{stack0}, \text{stack1}, \text{stack2}, \dots\} \leq \{\text{dataIn}, \text{stack0}, \text{stack1}, \dots\},$
- *acc: accumulate dataIn*
 $\{\text{stack0}, \text{stack1}, \text{stack2}, \dots\} \leq \{(\text{stack0} + \text{dataIn}), \text{stack1}, \text{stack2}, \dots\},$
- *swp: swap the last two recordings in stack*
 $\{\text{stack0}, \text{stack1}, \text{stack2}, \dots\} \leq \{\text{stack1}, \text{stack0}, \text{stack2}, \dots\}$
- *nop: no operation*
 $\{\text{stack0}, \text{stack1}, \text{stack2}, \dots\} \leq \{\text{stack0}, \text{stack1}, \text{stack2}, \dots\}.$

All the register buffered external connections are the following:

- $\text{dataIn}[31:0]$: data input provided by the external subsystem
- $\text{dataOut}[31:0]$: data output sent from the top of stack to the external subsystem
- $\text{aluCom}[3:0]$: command code executed by the unit
- carryIn : carry input
- carryOut : carry output

- `eqFlag` : is one if (`stack0 == stack1`)
- `ltFlag` : is one if (`stack0 < stack1`)
- `zeroFlag` : is one if (`stack0 == 0`)

Project 4.4

Chapter 5

PROCESSORS: Third order, 3-loop digital systems

*The soft overcomes the hard in the world
as a gentle rider controls a galloping horse.*

Lao Tzu¹

*The third loop allows the softness of symbols to act im-
posing the system's function.*

In order to add more autonomy in digital systems the third loop must be closed. Thus, new effects of the autonomy are used in order to reduce the complexity of the system. One of them will allow us to reduce the *apparent complexity* of an automaton, another, to reduce the complexity of the sequence of commands, but, the main form of manifesting of this third loop will be the *control process*.

The third loop can be closed in three manners, using the three types of circuits presented in the previous chapters.

- The first 3-OS type system is a system having the third loop closed through a *combinational circuit*, i.e., over an automaton or a network of automata the loop is closed through a 0-OS (see Figure 5.1a).
- The second type (see Figure 5.1b) has on the loop a *memory circuit* (1-OS).
- The third type connects in a loop two automata (see Figure 5.1c). This last type is typical for 3-OS, having the *processor* as the main component.

All these types of loops will be exemplified emphasizing a new and very important process appearing at the level of the third order system: **the segregation of the simple from the complex in order to reduce the global (apparent) complexity.**

¹Quote from *Tao Te King* of Lao Tzu translated by Brian Browne Walker.

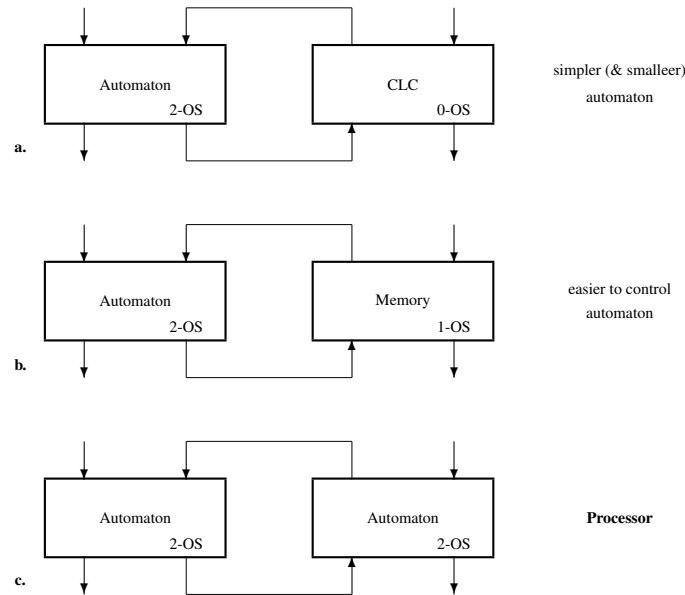


Figure 5.1: **The three types of 3-OS machines.** **a.** The third loop is closed through a combinational circuit resulting less complex, sometimes smaller, finite automaton. **b.** The third loop is closed through memories allowing a simplest control. **c.** The third loop is closed through another automaton resulting the **Processor**: the most complex and powerful circuit.

5.1 Automata using counters as registers

Are there ways to “extract” more “simplicity” by *segregation* from the PLA associated to an automaton? For some particular problems there is at least one more solution: to use a synchronous setable **counter**, $SCOUNT_n$. The synchronous setable counter is a circuit that combines two functions, it is a register (loaded on the command L) and in the same time it is a counter (counting up under the command U). The *load* has priority before the *count*.

Instead of using few one-bit counters, i.e. JK flip-flops, one few-bit counter is used to store the state and to simplify, *if possible*, the control of the state transition. The coding style used is the incremental encoding (see E.4.3), which provides the possibility that some state transitions to be performed by counting (increment).

Warning: *using setable counters is not always an efficient solution!*

Follows two example. One is extremely encouraging, and another is more realistic.

Example 5.1 *The half-automaton associated to the codes assignment written in parenthesis in Figure ?? is implemented using an $SCOUNT_n$ with $n = 2$. Because the states are coded using increment encoding, the state transitions in the flow-chart can be interpreted as follows:*

- *in the state q_0 if empty = 0, then the state code is incremented, else it remains the same*
- *in the state q_1 if empty = 0, then the state code is incremented, else it remains the same*
- *in the state q_2 if done = 1, then the state code is incremented, else it remains the same*

- in the state q_3 if $full = 0$, then the state code is incremented, else it remains the same

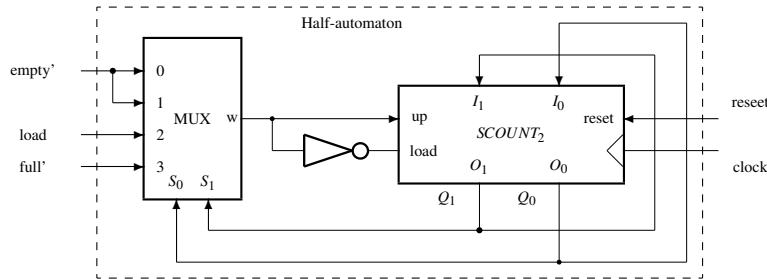


Figure 5.2: **Finite half-automaton implemented with a setable counter.** The last implementation of the half-automaton associated with FA from Figure ?? (with the function defined in Figure ?? where the states coded in parenthesis). A synchronous two-bit counter is used as state register. The simple four-input MUX commands the counter.

Results the very simple (not necessarily very small) implementation represented in Figure 5.2, where a 4-input multiplexer selects according to the state the way the counter switches: by increment ($up = 1$) or by loading ($load = 1$).

Comparing with the half-automaton part in the circuit represented in Figure ??, the version with counter is simpler, eventually smaller. But, the most important effect is the reducing complexity. \diamond

5.2 Loops closed through memories

Because the storage elements do not perform logical or arithmetical functions - they only store - a loop closed through the 1-OS seems to be unuseful or at least strange. But a selective memorizing action is used sometimes to optimize the computational process! The key is to know what can be useful in the next steps.

The previous two examples of the third order systems belongs to the subclass having a combinational loop. The function performed remains the same, only the efficiency is affected. In this section, because automata having the loop closed through a *memory* is presented, we expect the occurrence of some supplementary effects.

In order to exemplify how a trough memory loop works an *Arithmetic & Logic Automaton* – ALA – will be used (see Figure 5.3a). This circuit performs logic and arithmetic functions on data stored in its own state register called accumulator – ACC –, used as left operand and on the data received on its input *in*, used as right operand. A first version uses a **control** automaton to send commands to ALA, receiving back one flag: *crout*.

A second version of the system contains an additional D flip-flop used to store the value of the CR_{out} signal, in each clock cycle when it is enabled ($E = 1$), in order to be applied on the CR_{in} input of ALU. The control automaton is now substituted with a **command** automaton, used only to issue commands, without receiving back any flag.

Follow two example of using this ALA, one without an additional loop and another with the third loop closed trough a simple D flip-flop.

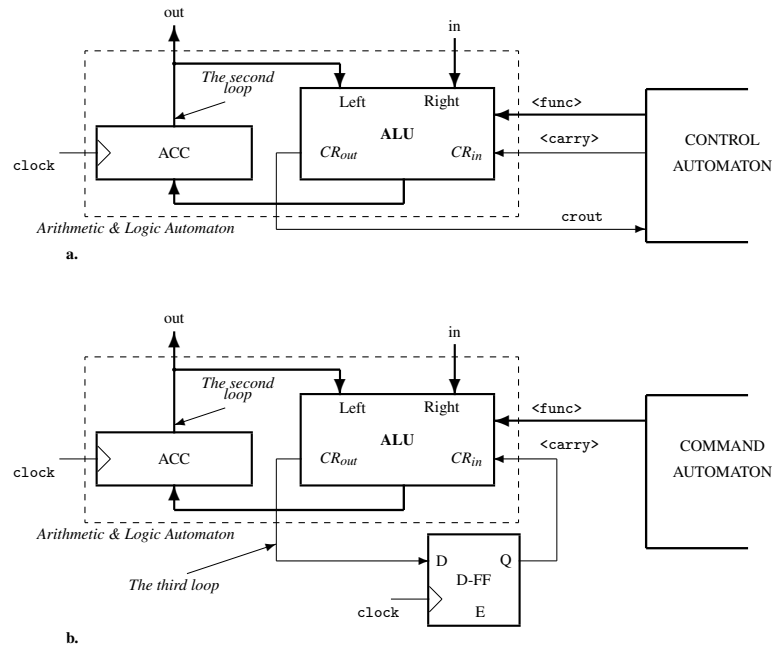


Figure 5.3: **The third loop closed over an arithmetic and logic automaton.** **a.** The basic structure: a simple automaton (its loop is closed through a simple combinational circuit: ALU) working under the supervision of a control automaton. **b.** The improved version, with an additional 1-bit state register to store the carry signal. The control is simpler if the third loop “tells” back to the arithmetic automaton the value of the carry signal in the previous cycle.

Version 1: the controlled Arithmetic & Logic Automaton

In the first case ALA is **controlled** (see Figure 5.3a) using the following definition for the undefined fields of `< microinstruction>` specified in 8.4.3:

```
<command> ::= <func> <carry>;
<func> ::= and | or | xor | add | sub | inc | shl | right;
<test> ::= crout | -;
```

Let be the sequence of commands that controls the increment of a double-length number:

```
inc cjmp crout bubu // ACC = in + 1
right jmp cucu     // ACC = in
bubu  inc          // ACC = in + 1
cucu  ...
```

The first increment command is followed by different operation according to the value of `crout`. If `crout = 1` then the next command is an increment, else the next command is a simple load of the upper bits of the double-length operand into the accumulator. The control automaton decides according to the result of the first increment and behaves accordingly.

Version 2: the commanded Arithmetic & Logic Automaton

The second version of *Arithmetic & Logic Automaton* is a 3-OS because of the additional loop closed through the D flip-flop. The role of this new loop is to reduce, to simplify and to speed up the routine that performs the same operation. Now the microinstruction is actualized differently:

```
<command> ::= <func>;
<func> ::= right | and | or | xor | add |
          sub | inc | shl | addcr | subcr | inccr | shlcr;
<test> ::= - ;
```

The field <test> is not used, and the control automaton can be substituted by a command automaton. The field <func> is coded so as one of its bit is 1 for all arithmetic functions. This bit is used to enable the switch of D-FF. New functions are added: *addcr*, *subcr*, *inccr*, *shlcr*. The instructions *xxxcr* operates with the value of carry F-F. The set of operations are defined now on *in*, *ACC*, *carry* with values in *carry*, *ACC*, as follows:

```
right: {carry, ACC} <= {carry, in}
and:   {carry, ACC} <= {carry, ACC & in}
or:    {carry, ACC} <= {carry, ACC | in}
xor:   {carry, ACC} <= {carry, ACC ^ in}
add:   {carry, ACC} <= ACC + in
sub:   {carry, ACC} <= ACC - in
inc:   {carry, ACC} <= in + 1
shl:   {carry, ACC} <= {in, 0}
addcr: {carry, ACC} <= ACC + in + carry
subcr: {carry, ACC} <= ACC - in - carry
inccr: {carry, ACC} <= in + carry
shlcr: {carry, ACC} <= {in, carry}
```

The resulting difference in how the system works is that in each clock cycle CR_{in} is given by the content of the D flip-flop. Thus, the sequence of commands that performs the same action becomes:

```
inc    // ACC = in + 1
inccr  // ACC = in + Q
```

In the two previous use of the arithmetic and logic automaton the execution time remains the same, but the expression used to command the structure in the second version is shorter and simpler. The explanation for this effect is the improved autonomy of the second version of the ALA. The first version was a 2-OS but the second version is a 3-OS. A significant part of the random content of the ROM from CROM can be removed by this simple new loop. Again, **more autonomy means less control**. A small circuit added as a new loop can save much from the random part of the structure. Therefore, this kind of loop acts as a *segregation method*.

Specific for this type of loop is that adding simple circuits we save random, i.e., complex, structured symbolic structures. The circuits grow by simple physical structure and the complex symbolic structures are partially avoided.

In the first version the sequence of commands are executed by the automaton all the time in the same manner. In the second version, a simpler sequence of commands are executed different, according to the processed data that impose different values in the carry flop-flop. This “different execution” can be thought as an “interpretation”.

In fact, the *execution* is substituted by the *interpretation*, so as the *apparent complexity* of the symbolic structure is reduced based on the additional autonomy due to the third structural loop. The autonomy introduced by the new loop through the D flip-flop allowed the interpretation of the commands received from the sequencer, according to the value of CR.

The third loop allows the simplest form of interpretation, we will call it *static interpretation*. The fourth loop allows a *dynamic interpretation*, as we will see in the next chapter.

5.3 Processors

Third-order systems are mainly represented by processing circuits. By processing we understand the modification of the symbolic structure of data in a way described by the symbolic structure of programs. Both data and programs are stored in RAM memories. The processing involves the following operations:

FETCH : reading from memory the current instruction

OPERATION : operation according to the current instruction which may consist of

- changing the internal state of the processor
 - using only the processor's internal salt
 - data fetched from memory
- modification of the data contained in the memory

NEXT : calculating the address of the next instruction

In the last 80 years, two extreme forms of implementing a processor have been imposed:

- the CISC processor (Complex Instruction Set Computer)
- the RISC processor (Reduction Instruction Set Computer)

derived from two abstract models configured in the 1940s:

- the von Neumann abstract model
- the Harvard abstract model

The distinction between these models materialized in hardware structures that were differentiated into two clearly distinct categories:

- processors that operated on the data by interpreting the instructions, which involves decomposing each instruction, usually complex, from the program into a sequence of microinstructions
- processors that operate on data by executing instructions, which involves the operation of a simple instruction in a single clock cycle

In the set of instructions of a RISC processor, only those instructions will be found that are simple, frequent and allow the realization of any instruction from the set of a CISC processor through a sequence of instructions.

5.3.1 Interpretive Processor: CISC Processor (RALU & CROM)

Generic structure of a CISC processor is represented in Figure 5.4, where:

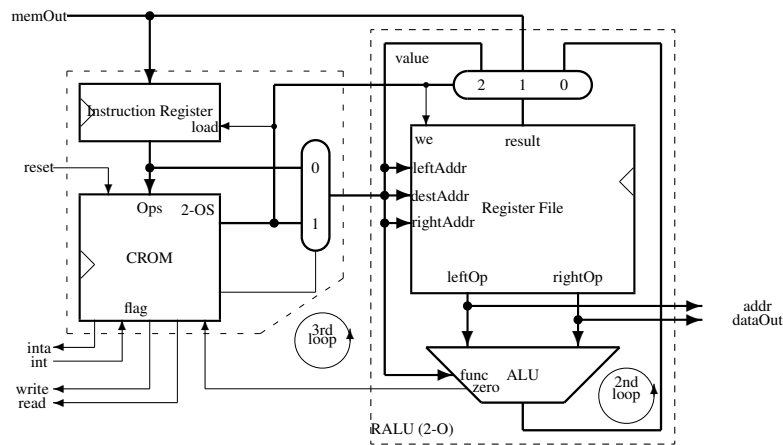


Figure 5.4: Hardware structure of a simple CISC.

Instruction Register : stores the current instruction during the interpretation which is a multi-cycle process

RALU : plays a role similar to the one in the EP structure (see ??); additionally has the role of memorizing the address of the subroutine run as a result of accepting the interrupt signal (*inta*), to which it adds a register *flag* for saving the return address from the subroutine

CROM : controls all the stages in the interpretation process by generating the enable signals for Instruction Register, Register File, and external memory; the *inta* signal is managed according to its internal state (in some cases a *inta* can be managed by a small and simple additional automaton serially connected in CROM.)

Multiplexor : allows, under the control of CROM, to use the command fields directly from Instruction Register or generated by CROM

The too high complexity that the CROM units reached in the economy of a processor and the statistics regarding the frequency with which the complex instructions were operated, led to the abandonment of this path of evolution. But history sometimes has surprising cycles. So the CISC approach also deserves a little attention.

5.3.2 Executive Processor: RISC Processor (RALU & PC)

Generic structure of a RISC processor is represented in Figure 5.5, where:

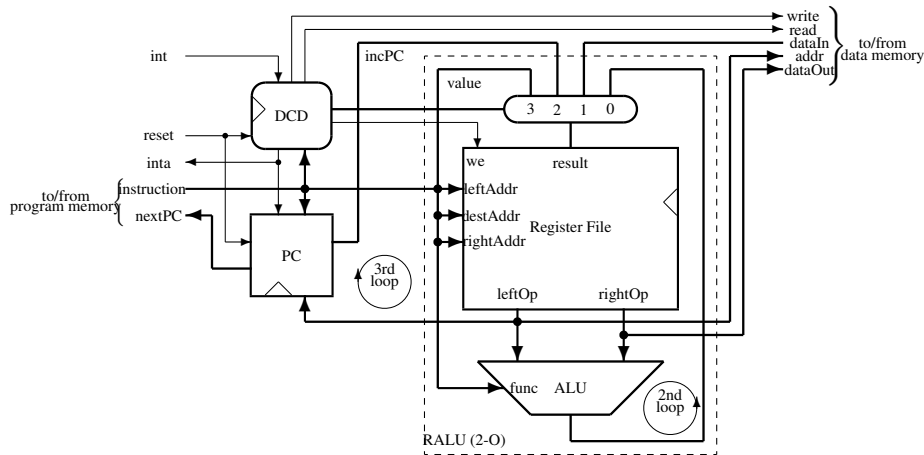


Figure 5.5: Hardware structure used for defining the architecture of a simple RISC.

RALU : plays a role similar to the one in the EP structure (see ??), but we can give up locating the PC here, for which we promote a specific structure

PC : it operates on the PC in parallel with the RALU that operates on the data content of the Register File, a fact that allows the execution of each instruction in a single cycle (of course, if we have separate and fast enough memories for program and data)

DCD : it is a complex circuit but of very small size if we design an efficient coding of the instructions

The structure of the RISC processor is very simple, because the size of the DCD is insignificant. Instead of the very complex and large combinational circuit (or ROM) in CROM, we now have a simple circuit. From the size point of view, we can afford to increase the size of some resources because they are simple. For example, we can have a larger Register File or/and an ALU with much more and more complex functions.

Next, we will delve deeper into issues related to RISC processors.

5.4 Case Study: *toyRISC* Processor

We will present in detail the RISC version of the processor through a case study of a simple but elaborate enough structure to illustrate the processor concept. We call *toyRISC* the processor that we will define, design and simulate.

5.4.1 The Concept of Processor's Architecture

We will define the concept of architecture with a small historical introduction. At the beginning of the 1960s, the company IBM (International Business Machine) had already launched several computer versions on the market, enough to highlight an unpleasant effect: for each new computer, the entire software development had to be reconsidered due to the hardware structure that justifiably suffered major

changes. As a consequence, on the occasion of the launch of a new series in [?] [?], the concept of the *architecture of a computing system* is proposed.

By the architecture of a processor we understand the *structural resources* by which the internal state of the processor is defined and the *set of instructions* by which this internal state evolves. Nothing about the way the structural resources are designed, or about their performance. The internal structure and its performance are at the discretion of the hardware designer. On the other hand, the way in which the set of instructions is used by the software designers is not the object of the architectural definition.

The architecture is, consequently, an interface between the hard and soft designers so that for long periods of time (in which several versions of the hard can be implemented) the already written software can be run on any new hard version. The assumed high cost of software development imposed this "inheritance" mechanism.

The architectural approach proves very useful for a limited number of hardware generations, but becomes a burden when technological conditions and market requirements change significantly.

The representation from Figure 5.5 represents the first stage in defining the architecture of a simple RISC processor, let's call it *toyRISC*. The next stage will be the micro-architectural definition used to specify the one-cycle micro-operations performed by each blocks. Finally, the Instruction Set Architecture (ISA) defined the instructions used to assembly the programs loaded in the program memory of the system.

5.4.2 toyRISC Micro-architecture

The micro-architecture of the processor toyRISC exemplified in Figure 5.5 is defined by the following storage resources:

pc [15:0] : the register used as program counter

ei : the state of an automaton used to enable the action of the interrupt signal int

rf [0:31] [31:0] : the register file

while in Figure ?? there is the file `DEFINES.vh` with the micro-operations executed by PC (the control operations), RALU (the arithmetic and logic operations) and DCD (memory transfer commands). The defined micro-operations represent only a part of what can be defined on the physical support provided by the structure represented in Figure 5.5. The reader can add additional operations using the following defined mechanisms.

```

/* *****
File name: DEFINES.vh
                MICROARCHITECTURE
***** */
// CONTROL
#define nop      6'b00_0000 // no operation: pc<=pc+1;
#define rjmp     6'b00_0001 // relative jump: pc<=pc+v;
#define zbr      6'b00_0010 // pc<=(rf[l]=0) ? pc+v:pc+1
#define nzbr     6'b00_0011 // pc<=!(rf[l]=0) ? pc+v:pc+1
#define ret      6'b00_0101 // return: pc<=rf[l][15:0];
#define halt     6'b00_0110 // halt until interrupt

```

```

`define eint      6'b00_1000 // set enable interrupt
`define dint      6'b00_1001 // set disable interrupt
// ARITHMETIC & LOGIC, for these instructions: pc<=pc+1;
`define add       6'b11_0000 // rf[d]<=rf[l]+rf[r];
`define sub       6'b11_0001 // rf[d]<=rf[l]-rf[r];
`define addv      6'b11_0010 // rf[d]<=rf[l]+v;
`define mult      6'b11_0011 // rf[d]<=rf[l]*rf[r];
`define multv    6'b11_0100 // rf[d]<=rf[l]*v;
`define addc      6'b11_0101 // rf[d]<=(rf[l]+rf[r])[32];
`define subc      6'b11_0110 // rf[d]<=(rf[l]-rf[r])[32];
`define addvc     6'b11_0111 // rf[d]<=(rf[l]+v)[32];
`define lsh       6'b11_1000 // rf[d]<=rf[l] >> 1;
`define ash       6'b11_1001 // rf[d]<=
                        // <={rf[l][31], rf[l][31:1]};
`define move      6'b11_1010 // rf[d]<=rf[l];
`define swap      6'b11_1011 // rf[d]<=
                        // <={rf[l][15:0], rf[l][31:16]};
`define bwnot     6'b11_1100 // rf[d]<=~rf[l];
`define bwand     6'b11_1101 // rf[d]<=rf[l]&rf[r];
`define bwor      6'b11_1110 // rf[d]<=rf[l]|rf[r];
`define bwxor     6'b11_1111 // rf[d]<=rf[l]^rf[r];
// MEMORY, for these instructions: pc=pc+1;
`define read      6'b10_0000 // read from dataMemory[rf[l]];
`define load      6'b10_0111 // rf[d]<=dataOut;
`define store     6'b10_1000 // dataMemory[rf[l]<=rf[r];
`define val       6'b01_0111 // rf[d]<={16*{v[15]},v};

```

The signal reset acts in PC ($pc \leq -1$) and in DCD by initializing the enable interrupt automaton ei ($ei \leq 0$, which means disable). Nothing in RALU is submitted to the initialization.

The PC block, controls the evolution of the program counter register pc . The content of this register evolves:

- by increment with 1 on the linear part of the program
- by increment with the immediate value provided by the instruction code
- by set to a value provided by the content of a register from the register file
- by set to a value provided by the instruction code (can be used to expand the set of operations already defined in Figure ??).

The RALU block, is mainly under the direct and full control of the instruction code $instr$ provided directly from the output of the program memory. Onlu the write back signal we is provided by DCD.

There are two instruction formats:

$$instr[31:0] = \{opCode[5:0], d[4:0], l[4:0], r[4:0], noUse[10:0]\} \mid \{opCode[5:0], d[4:0], l[4:0], v[15:0]\}$$

where:

opCode : is the operation code which specifies three types of operations:

- control operations acting on:
 - the value of the program counter, PC, which can be incremented or set to values according to the jumps or branches executed in program unconditionally or conditionally (in our simple processor, the only condition tested is if the value of the left operand is zero)
 - the state of the interrupt automaton used to enable the action of the interrupt signal `intIn`
- arithmetic-logic operations modify the content the register file according to the operations performed by ALU
- data transfer operations modify the content of the register file loading the immediate data or the memory data; the content of the external data memory is modified according to address and data stored in register file

d : specifies the destination of result provided by the ALU

l : specifies the left operand of the current operation

r : specifies the right operand of the current operation

v : is the immediate value used as right operand in the current operation

The first format of instruction operate only with the content of registers, while the second operate with the content of registers and an immediate value provided in the code of instruction: `v`.

5.4.3 toyRISC Instruction Set Architecture

The micro-architecture generates ISA by associating each micro operation of the operands and the destination located in the file register. In Figure ?? is listed an initial form of ISA (it can be expanded by adding micro-operations in the file `DEFINE.hv`).

```

/* *****
      toyRISC 'S ARCHITECTURE
***** */
NOP          // no operation
RJMP(1b)     // relative jumpto label 'lb'
ZBR(1,1b)    // branch if rf[l]=zero at label 'lb'
NZBR(1,1b)   // branch if rf[l]!=zero at label 'lb'
RET(1)       // return from subroutine: pc<=rf[l]
HALT        // halt until interrupt is received, pc = pc
// for the following instructions: pc<=pc+1;
EINT        // set enable interrupt
DINT        // set disable interrupt
ADD(d,1,r)  // rf[d]<=rf[l]+rf[r];
SUB(d,1,r)  // rf[d]<=rf[l]-rf[r];
ADDV(d,1,v) // rf[d]<=rf[l]+v;
MULT(d,1,r) // rf[d]<=rf[l]*rf[r];
MULTV(d,1,v)// rf[d]<=rf[l]*v;

```

```

ADDC(d, l, r) // rf[d]<=(rf[l]+rf[r])[32];
SUBC(d, l, r) // rf[d]<=(rf[l]-rf[r])[32];
ADDVC(d, l, v) // rf[d]<=(rf[l]+v)[32];
LSH(d, l) // rf[d]<=rf[l] >> 1;
ASH(d, l) // rf[d]<={rf[l][31], rf[l][31:1]};
MOVE(d, l) // rf[d]<=rf[l];
SWAP(d, l) // rf[d]<={rf[l][15:0], rf[l][31:16]};
NOT(d, l) // rf[d]<=~rf[l];
AND(d, l, r) // rf[d]<=rf[l]&rf[r];
OR(d, l, r) // rf[d]<=rf[l]|rf[r];
XOR(d, l, r) // rf[d]<=rf[l]^rf[r];
READ(l) // read from dataMemory[rf[l]];
LOAD(d) // rf[d]<=dataOut;
STORE(l, r) // dataMemory[rf[l]]<=rf[r];
VAL(d, v) // rf[d]<={{16*{v[15]}}, v};

```

5.4.4 toyRISC Implementation

The toyRISC processor will be implemented using a behavioral description in what follows, to provide a first picture of how circuits and information "work together" to provide complex functionality using a hardware structure dominated by *simple* structures and a *complex* software program. Indeed, the majority of the physical structure is made up of RALU and PC which are structures made up of simple circuits, and the DCD is made up of some complexly configured circuits, in the sense that their definition is in the same range as their size. On the other hand, the program that uses the toyRISC processor is a complex binary configuration, in the sense that it does not allow a lossy compression that, alone, could provide a compact representation. The program is what it is: a complex binary configuration.

The advantage of the combination between simple circuits and complex programs is at the functional level. We can build large circuits, because they are simple and we can afford complex programs because their design is done in a flexible environment where the error is tolerable because it is easily corrected. It is not so easy to correct a circuit error. In this way, the functionality of digital systems can reach very high levels of complexity.

We intend to test the competence of the toyRISC processor ignoring, at this stage, the performance that does not represent the target we are pursuing. For performance, hardware and software techniques are applied that exceed the circuit level to which we limit ourselves in this book. The next level of performance is extensively addressed in [?] [?].

Behavioral description

Because the project that we describe below emphasizes only the functional aspects leaving aside the aspects related to the performance, the description used for the main blocks are behavioral.

toyRISC.sv file looks at it is structurally described because the three files included are associated to the three main blocks represented in Figure 5.5: DCD, PC, RALU.


```

/* *****
File name: toyRISC.sv
                toyRISC
***** */
    'include "DEFINES.vh"
module toyRISC( input      [31:0]  instr   ,
                output    [15:0]  nextPC  ,
                input      intIn    ,
                output     inta     ,
                input      [31:0]  dataIn  ,
                output     [31:0]  dataOut ,
                output     [15:0]  addr   ,
                output reg   read    ,
                output reg   write   ,
                input       reset   ,
                input       clock   );

    reg [15:0]  pc      ;
    reg      ei      ;
    reg [31:0]  rf [0:31];
    wire [5:0]  opCode ;
    wire [4:0]  d, l, r ; // dest, lft, right
    wire [31:0] v      ; // immediate value
    reg      we      ;
    reg [1:0]  muxSel ;
    assign opCode = instr[31:26] ;
    assign d      = instr[25:21] ;
    assign l      = instr[20:16] ;
    assign r      = instr[15:11] ;
    assign v      = {{16{instr[15]}} , instr[15:0]};
    'include "DCDtoyRISC.sv"
    'include "PCtoyRISC.sv"
    'include "RALUtoyRISC.sv"
endmodule

```

A design for a real product is designed more carefully in terms of speed. For example, there are some places where pipe registers are needed to increase the clock frequency. We are content in our approach to illustrate the processing function as an important turning point in the structural evolution of digital systems towards structure-information symbiosis. Only the functional *competence* of the mixture circuit-program is considered in our approach. The performance is minimally considered or completely ignored.

`DCDtoyRISC.sv` file is the first file that we include in the top module (see Figure ??) describes the behavior of the decoder. It contains the 2-state interrupt automaton and the circuit which decodes the signals sent to the data memory.

```

/* *****

```

```

File name: DCDtoyRISC.sv
                toyRISC 's DCD
*****
always @(posedge clock)
    if (reset)                ei <= 0 ;
    else begin if (opCode == 'eint) ei <= 1 ;
                if (opCode == 'dint) ei <= 0 ;
                if (intIn & ei)      ei <= 0 ;
    end

assign inta = intIn & ei;

always @(*)
    casex (opCode)
        'read      : read  = 1'b1      ;
        'store     : write = 1'b1      ;
        default    : {read, write} = 2'b0 ;
    endcase

```

The interrupt automaton is designed to manage the acceptance of the action of the interrupt signal. Initially, the automaton is set on the state `disable interrupt` ($ei = 0$), because the program decides when the interrupt can be accepted, not before the register `rf[31]` is loaded with the address where the program associated to the interrupt is loaded. When the interrupt is accepted ($inta = 1$), the automaton switched in the state `disable interrupt` protecting the program from the action of another interrupt before the current one does it work. At the end of the program launched by the interrupt the interrupt automaton can be switched in the `enable interrupt`.

Important note: if the `halt` instruction runs and the interrupt automaton is in the `disable state`, then the entire system is blocked and the only solution to enable its behavior is to reset it.

`PCoyRISC.sv file` is the second file included in the `toyRISC.sv` file. It describes a simple automaton, the automaton whose state register in the program counter, `pc`. The automaton is an initial one. It can be initialized by the `reset` signal in the state `-1` to allow the evolution immediately after the end of `reset` starting with the instruction stored at the address `0` in the program memory. The PC automaton can also be initialized with the value stored in `rf[1]`; mechanism that allows the return of the program execution from the execution of the program associated with the interruption.

```

/* *****
File name: PCtoyRISC.sv
                toyRISC 's PC
*****
always @(posedge clock) if (reset)    pc <= -1      ;
                        else if (inta) pc <= rf[31];
                        else          pc <= nextPC;

always @(*)
    case (opCode)
        'rjmp    : nextPC = pc + v      ;

```

```

'zbf      : nextPC = ( rf[1] == 0 ) ? pc + v : pc + 1 ;
'nzbf     : nextPC = ( rf[1] != 0 ) ? pc + v : pc + 1 ;
'ret      : nextPC = rf[1] ;
'halt     : nextPC = pc ;
default  : nextPC = pc + 1 ;

```

Otherwise, the automaton evolves depending on the state it is in, `pc`, the command received via `opCode` and depending on the value of the left operand which is tested if it is or not zero.

`RALUtoyRISC.sv` file is the third file included in the `toyRISC.sv` file. It describes also a simple automaton. Its structured state, stored in a memory organized as a register file, is submitted to the processing defined as the sequence of the arithmetic and logic operations performed by a simple circuit: ALU.

The signal `inta` it also acts here, as in the case of the PC, having priority over the operation code received from the program memory: register `rf[30]` takes the return value (`pc+1`) from the program (subroutine) run as a result of the interruption.

The `always` form describes a half-automaton with the input:

```
{instr, inta, dataIn, pc}
```

and the internal state as the following Cartesian product:

```
RF = {rf[0], rf[1], ..., rf[31]}
```

```

/*****
File name: RALUtoyRISC.sv
          toyRISC's RALU
*****/
always @(posedge clock)
  if (inta)      rf[30] <= pc + 1 ;
  else
    case (opCode)
      'add      : rf[d] <= rf[1]+rf[r] ;
      'sub      : rf[d] <= rf[1]-rf[r] ;
      'addv     : rf[d] <= rf[1]+v ;
      'mult     : rf[d] <= rf[1]*rf[r] ;
      'multv    : rf[d] <= rf[1]*v ;
      'addc     : rf[d] <= ((rf[1]+rf[r]) & 2^32 == 0)
                    ? 0 : 1 ;
      'subc     : rf[d] <= ((rf[1]-rf[r]) & 2^32 == 0)
                    ? 0 : 1 ;
      'addvc    : rf[d] <= ((rf[1]+v) & 2^32 == 0)
                    ? 0 : 1 ;
      'lsh      : rf[d] <= rf[1] >> 1 ;
      'ash      : rf[d] <= {rf[1][31], rf[1][31:1]} ;
      'move     : rf[d] <= rf[1] ;
      'swap     : rf[d] <= {rf[1][15:0], rf[1][31:16]} ;

```

```

        'bwnot : rf[d] <= ~rf[l] ;
        'bwand : rf[d] <= rf[l]&rf[r] ;
        'bwor  : rf[d] <= rf[l]|rf[r] ;
        'bwxor : rf[d] <= rf[l]^rf[r] ;
        'load  : rf[d] <= dataIn ;
        'val   : rf[d] <= v ;
        default : rf[0] <= rf[0] ;
    endcase
assign addr    = rf[l][9:0];
assign dataOut = rf[r] ;

```

The outputs associated to the half-automaton are described by the last two `assign` form which take the address and data for data memory from directly from the register file's output.

5.5 Concluding about the third loop

The third loop is closed through simple automata avoiding the fast increasing of the complexity in digital circuit domain. It allows the autonomy of the control mechanism.

"Intelligent registers" ask less structural control maintaining the complexity of a finite automaton at the smallest possible level. Intelligent, loop driven circuits can be controlled using smaller complex circuits.

The loop through a storage element ask less symbolic control at the micro-architectural level. Less symbols are used to determine the same behavior because the local loop through a memory element generates additional information about the recent history.

Looping through a memory circuit allows a more complex "understanding" because the controlled circuits "knows" more about its behavior in the previous clock cycle. The circuit is somehow "conscious" about what it did before, thus being more "responsible" for the operation it performs now.

Looping through an automaton allows any effective computation. Using the theory of computation (see chapter *Recursive Functions & Loops* in this book) can be proved that any effective computation can be done using a three loop digital system. More than three loops are needed only for improving the efficiency of the computational structures.

The third loop allows the symbolic functional control using the arbitrary meaning associated to the binary codes embodied in instructions or micro-instructions. Both, the coding and the decoding process being controlled at the design level, the binary symbols act actualizing the potential structure of a programmable machine.

Real processors use circuit level parallelism discussed in the first chapter of this book. They are: data parallelism, time parallelism and speculative parallelism. How all these kind of parallelism are used is a computer architecture topic, beyond the goal of these lecture notes.

5.6 Problems

Problem 5.1 *Interrupt automaton with asynchronous input.*

Problem 5.2 *Solving the second degree equations with an elementary processor.*

Problem 5.3 *Compute y if x , m and n is given with an elementary processor.*

Problem 5.4 *Modify the unending loop of the processor to avoid spending time in testing if a new instruction is in inFIFO when it is there.*

Problem 5.5 *Define an instruction set for the processor described in this chapter using its microarchitecture.*

Problem 5.6 *Our CISC Processor: how must be coded the instruction set to avoid FUNC MUX?*

5.7 Projects

Project 5.1 *Design a specialized elementary processor for rasterization function.*

Project 5.2 *Design a system integrating in a parallel computational structure 8 rasterization processors designed in the previous project.*

Project 5.3 *Design a floating point arithmetic coprocessor.*

Project 5.4 *Design the RISC processor defined by the following Verilog behavioral description:*

```
module risc_processor(
    );

endmodule
```

Project 5.5 *Design a version of Stack Processor modifying SALU as follows: move MUX4 to the output of ALU and the input of STACK.*

Chapter 6

COMPUTING MACHINES: ≥4-loop digital systems

Software is getting slower more rapidly than hardware becomes faster.

Wirth's law¹

To compensate the effects of the bad behavior of software guys, besides the job done by the Moore law a lot of architectural work must be added.

The last examples of the previous chapter emphasized a process that appears as a "turning point" in 3-OS: the function of the system becomes lesser and lesser dependent on the *physical structure* and the function is more and more assumed by a *symbolic structure* (the program or the microprogram). The physical structure (the circuit) remains simple, rather than the symbolic structure, "stored" in program memory of in a ROM, that establishes the functional complexity. The fourth loop creates the condition for a total functional dependence on the symbolic structure. By the rule, at this level an *universal circuit* - the **processor** - *executes* (in RISC machines) or *interprets* (in CISC machines) symbolic structures stored in an additional device: the *program memory*.

6.1 Types of fourth order systems

There are four main types of fourth order systems (see Figure 6.1) depending on the order of the system through which the loop is closed:

1. **P & ROM** is a 4-OS with loop closed through a 0-OS - in Figure 6.1a the combinational circuit is a ROM containing only the programs executed or interpreted by the processor

¹Niklaus Wirth is an already legendary Swiss born computer scientist with many contributions in developing various programming languages. The best known is Pascal. *Wirth's law* is a sentence which Wirth made popular, but he attributed it to Martin Reiser.

2. **P & RAM** is a 4-OS with loop closed through a 1-OS - is the **computer**, the most representative structure in this order, having on the loop a RAM (see Figure 6.1b) that stores both data and programs
3. **P & LIFO** is a 4-OS with loop closed through a 2-OS - in Figure 6.1c the automaton is represented by a push-down stack containing, by the rule, data (or sequences in which the distinction between data and programs does not make sense, as in the Lisp programming language, for example)
4. **P & CO-P** is a 4-OS with loop closed through a 3-OS - in Figure 6.1d COPROCESSOR is also a processor but a specialized one executing efficiently critical functions in the system (in most of cases the coprocessor is a floating point arithmetic processor).

The representative system in the class of **P & ROM** is the *microcontroller* the most successful circuit in 4-OS. The microcontroller is a “best seller” circuit realized as a one-chip computer. The core of a microcontroller is a processor executing/interpreting the programs stored in a ROM.

The representative structure in the class of **P & RAM** is the computer. More precisely, the structure *Processor - Channel - Memory* represents the physical support for the well known *von Neumann architecture*. Almost all present-day computers are based on this architecture.

The third type of system seems to be strange, but a recent developed architecture is a *stack oriented architecture* defined for the successful Java language. Naturally, a real Java machine is endowed also with the program memory.

The third and the fourth types are machines in which the segregation process emphasized physical structures, a stack or a coprocessor. In both cases the segregated structures are also simple. The consequence is that the whole system is also a simple system. But, the first two systems are very complex systems in which the simple is not segregated by the random. The support of the random part is the ROM *physical structure* in the first case and the *symbolic content* of the RAM memory in the second.

The actual computing machines have currently more than order 4, because the processors involved in the applications have additional features. Many of these features are introduced by new loops that increase the autonomy of certain subsystems. But theoretically, the computer function asks at least four loops.

6.2 The computer – support for the strongest segregation

The ROM content is defined symbolically and after that it is converted in the actual physical structure of ROM. Instead, the RAM content remains in symbolic form and has, in consequence, more flexibility. This is the main reason for considering the PROCESSOR & RAM = COMPUTER as the most representative in 4-OS.

The computer is not a circuit. It is a new entity with a special functional definition, currently called **computer architecture**. Mainly, the computer architecture is given by the machine language. A program written in this language is interpreted or executed by the processor. The program is stored in the RAM memory. In the same subsystem are stored data on which the program “acts”. Each architecture can have many associated computer structures (organizations).

Starting from the level of four order systems the behavior of the system is controlled mainly by the symbolic structure of programs. The architectural approach settles the distinction between the physical structures and the symbolic structures. Therefore, any computing **machine** supposes the following triadic definition (suggested by [”Milutinovic” ’89]):

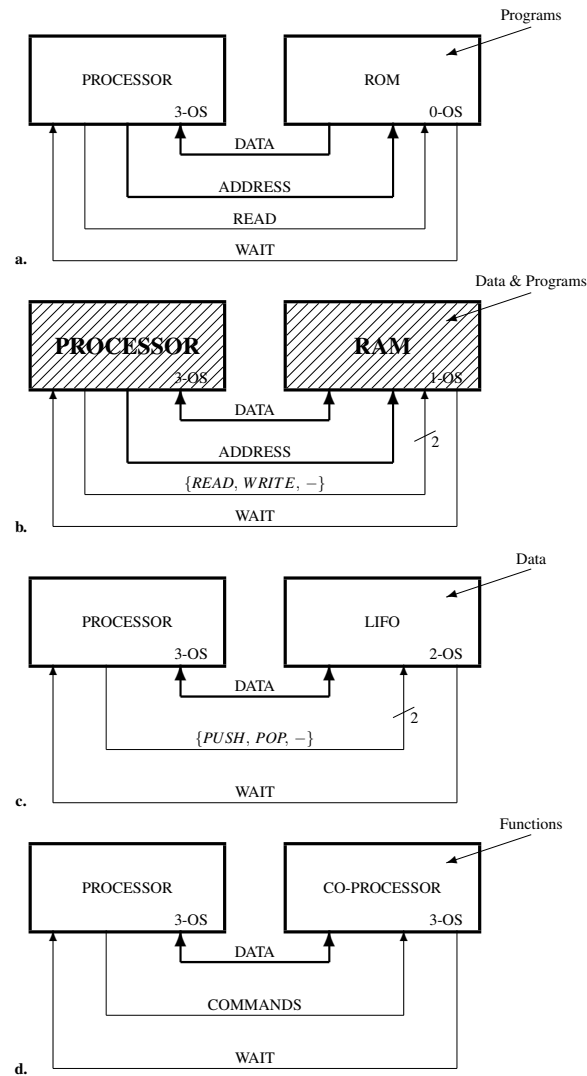


Figure 6.1: **The four types of 4-OS machines.** **a.** Fix program computers usual in embedded computation. **b.** General purpose computer. **c.** Specialized computer working on a restricted data structure. **d.** Accelerated computation supported by a specialized co-processor.

- the machine language (usually called *architecture*)
- the storage containing programs written in the machine language
- the **machine** that *interprets* the programs, containing:
 - the machine language ...
 - the storage ...
 - the **machine** ... containing:
 - * ...

and so on until the **machine** *executes* the programs.

Does it make any sense to add new loops? Yes, but not too much! It can be justified to add loops inside the processor structure to improve its capacity to interpret fast the machine language by using simple circuits. Another way is to see PROCESSOR & COPROCESSOR or PROCESSOR & LIFO as performant processors and to add over them the loop through RAM. But, mainly these machines remain structures having the computer function. The computer needs at least four loops to be *competent*, but currently it is implemented on system having more loops in order to become *performant*.

6.2.1 Four-Loop Circuits (4-OS) & Controlling by Information

In the previous subsection, the information interacts directly with the physical structure. All the information is executed or interpreted by the circuits. The next step disconnects partially the information from circuits. In a system, having four loops the information can be interpreted by another information acting to the lower level in the system. The typical 4-OS is the *computer* structure (see Chapter ??). This structure is more than we need for computing. Indeed, as we said in subsection ??, the partial recursive functions can be computed in 3-OSs. Why are we interested in using 4-OS for performing computations? The answer is: *for segregating more the simple circuits from random (complex) informational structure*. In a system having four loops the simple and the complex are maximal segregated, the first in circuits and the second in information.

At the 3-OS level, the information also interacts, and thereby acts, with the structure of the circuits through the flags. The control performed depends on what happens directly in the controlled circuits. At the 4-OS level, the control is taken over by the information in an imperative way, a way that no longer depends at all on the signals coming directly from the circuits.

Starting from the level of the fourth order systems the functional aspects of a digital system is imposed mainly by the information. The role of the circuits decreases. Circuits become simple even if they gain in size. The complexity of the computation switches from circuits to information.

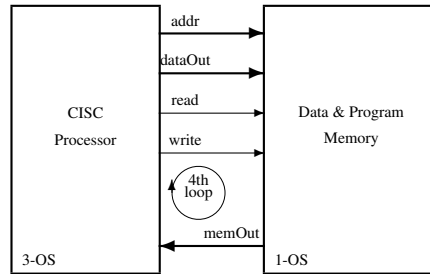


Figure 6.2: von Neumann abstract model for computer.

6.2.2 Five-Loop Circuits (5-OS): Computer with RISC Processor

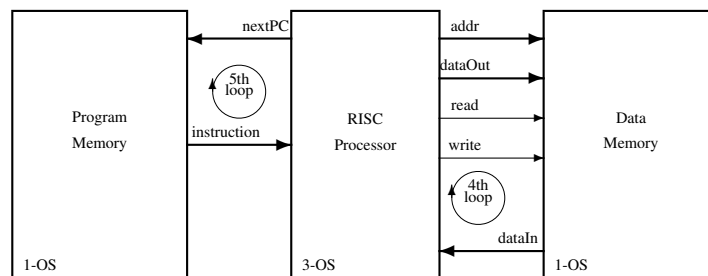


Figure 6.3:

6.3 Problems

Problem 6.1 *Interpretative processor with distinct program counter block.*

6.4 Projects

Project 6.1

ANNEXES

Appendix A

Binary Arithmetic

A.1 Binary representations

A.1.1 Positive integers

Positive integer denoted by \mathbb{Z}^+ , and they are the solution to the simple linear recurrence equation $a_n = a_{n-1} + 1$ with $a_1 = 1$.

A n -bit number:

$$B_{n-1}B_{n-2}\dots B_1B_0 \Rightarrow B_{n-1} \times 2^{n-1} + B_{n-2} \times 2^{n-2} + \dots + B_1 \times 2^1 + B_0 \times 2^0$$

where $B_i \in \{0, 1\}$ for $i = 0, 1, \dots, n-1$.

A.1.2 Decimal to binary conversion

The algorithm of converting the decimal number D in a n -bit binary form is:

step 1 $D_0 = \lfloor D/2 \rfloor$ // the whole part of D divided by 2
 $B_0 = D - \lfloor D/2 \rfloor \times 2$ // the remainder of dividing D by 2

step 2 $D_1 = \lfloor D_0/2 \rfloor$
 $B_1 = D_0 - \lfloor D_0/2 \rfloor \times 2$

...

step n $D_{n-1} = \lfloor D_{n-2}/2 \rfloor$
 $B_{n-1} = D_{n-2} - \lfloor D_{n-2}/2 \rfloor \times 2$

A.1.3 Signed integers

Sign-magnitude representation

{sign, magnitude}

0_0000 => +0

0_0001 => +1

0_0010 => +2

```

...
0_1111 => +15
1_0000 => -0
1_0001 => -1
1_0010 => -2
...
1_1111 => -15

```

Ones' complement representation

Ones' complement is bitwise negation.

```

{sign, magnitude}

0_0000 => +0
0_0001 => +1
0_0010 => +2
...
0_1111 => +15
1_0000 => -15
1_0001 => -14
1_0010 => -13
...
1_1111 => -0

```

Two's complement representation

Two's complement is ones' complement plus 1

```

{sign, magnitude}

0_0000 => +0
0_0001 => +1
0_0010 => +2
...
0_1111 => +15
1_0000 => -16
1_0001 => -15
1_0010 => -14
...
1_1111 => -1

```

A.1.4 Fix point fractionary numbers

$$B_{n-1} \dots B_1 B_0 . F_1 F_2 \dots \Rightarrow B_{n-1} \times 2^{n-1} + \dots + B_1 \times 2^1 + B_0 \times 2^0 + F_1 \times 2^{-1} + F_2 \times 2^{-2} + \dots$$

where $B_i, F_i \in \{0, 1\}$ for $i = 0, 1, \dots$

A.1.5 Floating point numbers`{sign, exponent, fraction}`

$$\text{sgn } 1.\text{fraction} \times 2^{\text{exponent}-127}$$

IEEE half-precision 16-bit float: deep learning artificial intelligence`{sign, exponent[4:0], fraction[9:0]}`**Google's brain float** bfloat16 is a 16-bit float:`{sign, exponent[7:0], fraction[6:0]}`**NVidia's TensorFloat** 19-bit float:`{sign, exponent[7:0], fraction[9:0]}`**AMD's fp24** 24-bit float:`{sign, exponent[6:0], fraction[14:0]}`**Pixar's PXR24** 24-bit float:`{sign, exponent[7:0], fraction[14:0]}`**IEEE 754 single-precision** 16-bit float:`{sign, exponent[7:0], fraction[22:0]}`**Example A.1** In IEEE 754 single-precision, the number:`0_10000010_110000000000000000000000`

corresponds to:

$$+(1 + (0.5 + 0.25)) \times 2^{130-127} = 14$$

◇

A.2 Adding/Subtracting

The most efficient representation for add/sub is twos complement.

A.2.1 Adding positive integers

Carry represents the overflow for positive integer addition. We consider integers represented on 4 bits.

A first example with no carry:

```
+1      0001
+6      0110
--      ----
+5      0111
```

An example with carry:

```
+10     1010
+8      1000
--      ----
(1)+16  (1)0010
```

A.2.2 Adding signed integers

Simply add the numbers and ignore any carry out of the highest bit.

```
-1      1_1111
+6      0_0110
--      -----
+5      (1)0_0101

+1      0_0001
-4      1_1100
--      -----
-3      (0)1_1101
```

A.2.3 Subtracting

A-B means A+(2s compl of B)

$$3-2 \Rightarrow 0_0011 + \text{NOT}(0_0010) + 1 \Rightarrow 0_0011 + 1_1101 + 1 \Rightarrow (1)0_0001 = 0_0001$$

A.2.4 Overflow

In signs of the operands, sgn1 and sgn2 , are different, overflow is not possible. If they are the same overflow is possible if the sign of the result, sgnR , is different from the common signs of operands.

$$\text{overflow} = (\text{sgn1} \wedge \text{sgn0})' \& (\text{sgn1} \wedge \text{sgnR})$$

A.3 Multiply/Divide

A.3.1

A.3.2

Appendix B

Boolean functions

Searching the truth, dealing with numbers and behaving automatically are all based on logic. Starting from the very elementary level we will see that logic can be “interpreted” arithmetically. We intend to offer a physical support for both the numerical functions and logical mechanisms. The logic circuit is the fundamental brick used to build the physical computational structures.

B.1 Short History

There are some significant historical steps on the way from logic to numerical circuits. In the following some of them are pointed.

Aristotle of Stagira (382-322) a Greek philosopher considered as founder for many scientific domains. Among them logics. All his writings in logic are grouped under the name *Organon*, that means *instrument* of scientific investigation. He worked with two logic values: **true** and **false**.

George Boole (1815-1864) is an English mathematician who formalized the Aristotelian logic like an algebra. The *algebraic logic* he proposed in 1854, now called *Boolean logic*, deals with the truth and the false of complex expressions of binary variables.

Claude Elwood Shannon (1916-2001) obtained a master degree in electrical engineering and PhD in mathematics at MIT. His Master’s thesis, *A Symbolic Analysis of Relay and Switching Circuits* [Shannon ’38], used Boolean logic to establish a theoretical background of digital circuits.

B.2 Elementary circuits: gates

Definition B.1 A **binary variable** takes values in the set $\{0, 1\}$. We call it *bit*.

The set of numbers $\{0, 1\}$ is interpreted in logic using the correspondences: $0 \rightarrow \text{false}$, $1 \rightarrow \text{true}$ in what is called *positive logic*, or $1 \rightarrow \text{false}$, $0 \rightarrow \text{true}$ in what is called *negative logic*. In the following we use positive logic.

Definition B.2 We call **n -bit binary variable** an element of the set $\{0, 1\}^n$.

Definition B.3 A **logic function** is a function having the form $f : \{0, 1\}^n \rightarrow \{0, 1\}^m$ with $n \geq 0$ and $m > 0$.

In the following we will deal with $m = 1$. The parallel composition will provide the possibility to build systems with $m > 1$.

B.2.1 Zero-input logic circuits

Definition B.4 The **0-bit logic function** are $f_0^0 = 0$ (the false-function) which generates the one bit coded 0, and $f_1^0 = 1$ (the true-function) which generate the one bit coded 1.

They are useful for generating initial values in computation (see the *zero* function as basic function in partial recursivity).

B.2.2 One input logic circuits

Definition B.5 The **1-bit logic functions**, represented by **true-tables** in Figure B.1, are:

- $f_0^1(x) = 0$ – the false function
- $f_1^1(x) = x'$ – the invert (not) function
- $f_2^1(x) = x$ – the driver or identity function
- $f_3^1(x) = 1$ – the true function

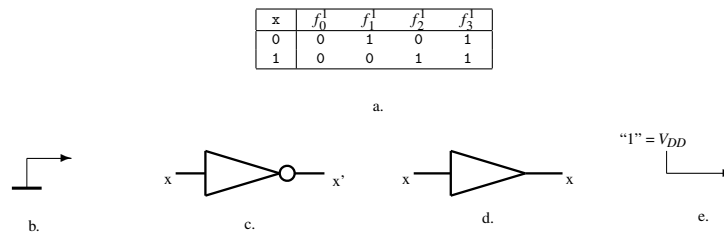


Figure B.1: **One-bit logic functions.** **a.** The truth table for 1-variable logic functions. **b.** The circuit for “0” (false) by connecting to the ground potential. **c.** The logic symbol for the inverter circuit. **d.** The logic symbol for driver function. **e.** The circuit for “1” (true) by connecting to the high potential.

Numerical interpretation of the NOT circuit: **one-bit incrementer**. Indeed, the output represents the modulo 2 increment of the inputs.

B.2.3 Two inputs logic circuits

Definition B.6 The **2-bit logic functions** are represented by true-tables in Figure B.2.

Interpretations for some of 2-input logic circuits:

- f_8^2 : AND function is:
 - a multiplier for 1-bit numbers
 - a **gate**, because x opens the gate for y :
if ($x = 1$) output = y ; **else** output = 0;
- f_6^2 : XOR (exclusiv OR) is:

x y	f_0^2	f_1^2	f_2^2	f_3^2	f_4^2	f_5^2	f_6^2	f_7^2	f_8^2	f_9^2	f_{10}^2	f_{11}^2	f_{12}^2	f_{13}^2	f_{14}^2	f_{15}^2
0 0	0	1	0	1	0	1	0	1	0	1	0	1	0	1	0	1
0 1	0	0	1	1	0	0	1	1	0	0	1	1	0	0	1	1
1 0	0	0	0	0	1	1	1	1	0	0	0	0	1	1	1	1
1 1	0	0	0	0	0	0	0	0	1	1	1	1	1	1	1	1

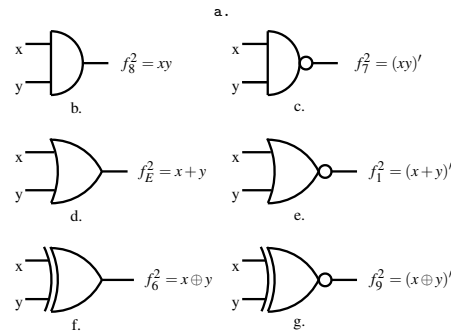


Figure B.2: **Two-bit logic functions.** **a.** The table of all two-bit logic functions. **b.** AND gate – the original gate. **c.** NAND gate – the most used gate. **d.** OR gate. **e.** NOR gate. **f.** XOR gate – modulo2 adder. **g.** NXOR gate – coincidence circuit.

- the 2-modulo adder
- NEQ (not-equal) circuit, a comparator pointing out when the two 1-bit numbers on the input are unequal
- an enabled inverter:
 - if** $x = 1$ output is y' ; **else** output is y ;
- a modulo 2 incrementer.
- f_B^2 : the logic implication is also used to compare 1-bit numbers because the output is 1 for $y < x$
- f_1^2 : NOR function detects when 2-bit numbers have the value zero.

All logic circuits are *gates*, even if a true gate is only the AND *gate*.

B.2.4 Many input logic circuits

For enumerating the 3-input function a table with 8 line is needed. On the left side there are 3 columns and on the right side 256 columns (one for each 8-bit binary configuration defining a logic function).

Theorem B.1 *The number of n-input one output logic (Boolean) functions is $N = 2^{2^n}$.* \diamond

Enumerating is not a solution starting with $n = 3$. Maybe the 3-input function can be defined using the 2-input functions.

B.3 How to Deal with Logic Functions

The systematic and formal development of the **theory** of logical functions means: (1) a set of elementary functions, (2) a minimal set of axioms (of formulas considered true), and (3) some rule of deduction.

Because our approach is a **pragmatic** one: (1) we use an extended (non-minimal) set of elementary functions containing: NOT, AND, OR, XOR (a minimal one contains only NAND, or only NOR), and (2) we will list a set of useful principles, i.e., a set of **equivalences**.

Identity principle Even if the natural tendency of existence is becoming, we stone the value a to be identical with itself: $a = a$. Here is one of the fundamental limits of digital systems and of computation based on them.

Double negation principle The negation is a “reversible” function, i.e., if we know the output we can deduce the input (it is a very rare, somehow unique, feature in the world of logical function): $(a')' = a$. Actually, we can not found the reversibility in existence. There are logics that don’t accept this principle (see the intuitionist logic of Heyting & Brouwer).

Associativity Having 2-input gates, how can be built gates with much more inputs? For some functions the associativity helps us.

$$a + (b + c) = (a + b) + c = a + b + c$$

$$a(bc) = (ab)c = abc$$

$$a \oplus (b \oplus c) = (a \oplus b) \oplus c = a \oplus b \oplus c.$$

Commutativity Commutativity allows us to connect to the inputs of **some** gates the variable in any order.

$$a + b = b + a$$

$$ab = ba$$

$$a \oplus b = b \oplus a$$

Distributivity Distributivity offers the possibility to define **all** logical functions as *sum of products* or as *product of sums*.

$$a(b + c) = ab + ac$$

$$a + bc = (a + b)(a + c)$$

$$a(b \oplus c) = ab \oplus ac.$$

Not all distributions are possible. For example:

$$a \oplus bc \neq (a \oplus b)(b \oplus c).$$

The table in Figure B.3 can be used to prove the previous inequality.

a	b	c	bc	$a \oplus bc$	$a \oplus b$	$a \oplus c$	$(a \oplus b)(a \oplus c)$
0	0	0	0	0	0	0	0
0	0	1	0	0	0	1	0
0	1	0	0	0	1	0	0
0	1	1	1	1	1	1	1
1	0	0	0	1	1	1	1
1	0	1	0	1	1	0	0
1	1	0	0	1	0	1	0
1	1	1	1	0	0	0	0

Figure B.3: **Proving by tables.** Proof of inequality $a \oplus bc \neq (a \oplus b)(b \oplus c)$.

Absorbtion Absorbtion simplify the logic expression.

$$a + a' = 1$$

$$a + a = a$$

$$aa' = 0$$

$$aa = a$$

$$a + ab = a$$

$$a(a + b) = a$$

$$\textit{Tertium non datur: } a + a' = 1.$$

Half-absorbtion The half-absorbtion allows only a smaller, but non-neglecting, simplification.

$$a + a'b = a + b$$

$$a(a' + b) = ab.$$

Substitution The substitution principles say us what happen when a variable is substituted with a value.

$$a + 0 = a$$

$$a + 1 = 1$$

$$a0 = 0$$

$$a1 = a$$

$$a \oplus 0 = a$$

$$a \oplus 1 = a'.$$

Exclusion The most powerful simplification occurs when the exclusion principle is applicable.

$$ab + a'b = b$$

$$(a + b)(a' + b) = b.$$

Proof. For the first form:

$$ab + a'b = b$$

applying successively distribution, absorbtion and substitution results:

$$ab + a'b = b(a + a') = b1 = b.$$

For the second form we have the following sequence:

$$(a + b)(a' + b) = (a + b)a' + (a + b)b = aa' + a'b + ab + bb =$$

$$0 + (a'b + ab + b) = a'b + ab + b = a'b + b = b.$$

De Morgan laws Some times we are interested to use inverting gates instead of non-inverting gates, or conversely. De Morgan laws will help us.

$$a + b = (a'b')' \quad ab = (a' + b')'$$

$$a' + b' = (ab)'\quad a'b' = (a + b)'$$

B.4 Minimizing Boolean functions

Minimizing logic functions is the first operation to be done after defining a logical function. Minimizing a logical function means to express it in the simplest form (with minimal symbols). To a simple form a small associated circuit is expected. The minimization process starts from canonical forms.

B.4.1 Canonical forms

The initial definition of a logic function is usually expressed in a canonical form. The canonical form is given by a truth table or by the rough expression extracted from it.

Definition B.7 A **minterm** associated to an n -input logic function is a logic product (AND logic function) depending by all n binary variable. \diamond

Definition B.8 A **maxterm** associated to an n -input logic function is a logic sum (OR logic function) depending by all n binary variable. \diamond

Definition B.9 The **disjunctive normal form**, DNF, of an n -input logic function is a logic sum of minterms. \diamond

Definition B.10 The **conjunctive normal form**, CNF, of an n -input logic function is a logic product of maxterms. \diamond

Example B.1 Let be the combinational multiplier for 2 2-bit numbers described in Figure B.4. One number is the 2-bit number $\{a, b\}$ and the other is $\{c, d\}$. The result is the 4-bit number $\{p_3, p_2, p_1, p_0\}$. The logic equations result direct as 4 DNFs, one for each output bit:

$$p_3 = abcd$$

$$p_2 = ab'cd' + ab'cd + abcd'$$

$$p_1 = a'bcd' + a'bcd + ab'c'd + ab'cd + abc'd + abcd'$$

$$p_0 = a'bc'd + a'bcd + abc'd + abcd.$$

Indeed, the p_3 bit takes the value 1 only if $a = 1$ and $b = 1$ and $c = 1$ and $d = 1$. The bit p_2 is 1 only one of the following three 4-input ADNs takes the value 1: $ab'cd'$, $ab'cd$, $abcd'$. And so on for the other bits.

Applying the De Morgan rule the equations become: $p_3 = ((abcd)')$

$$p_2 = ((ab'cd')'(ab'cd)'(abcd')')$$

$$p_1 = ((a'bcd')'(a'bcd)'(ab'c'd)'(ab'cd)'(abc'd)'(abcd')')$$

$$p_0 = ((a'bc'd)'(a'bcd)'(abc'd)'(abcd)')$$

These forms are more efficient in implementation because involve the same type of circuits (NANDs), and because the inverting circuits are usually faster.

The resulting circuit is represented in Figure B.5. It consists in two layers of ADNs. The first layer computes only minterms and the second “adds” the minterms thus computing the 4 outputs.

The logic depth of the circuit is 2. But in real implementation it can be bigger because of the fact that big input gates are composed from smaller ones. Maybe a real implementation has the depth 3. The propagation time is also influenced by the number of inputs and by the fan-out of the circuits.

The size of the resulting circuit is very big also: $S_{mult_2} = 54$. \diamond

ab	cd	p3	p2	p1	p0
00	00	0	0	0	0
00	01	0	0	0	0
00	10	0	0	0	0
00	11	0	0	0	0
01	00	0	0	0	0
01	01	0	0	0	1
01	10	0	0	1	0
01	11	0	0	1	1
10	00	0	0	0	0
10	01	0	0	1	0
10	10	0	1	0	0
10	11	0	1	1	0
11	00	0	0	0	0
11	01	0	0	1	1
11	10	0	1	1	0
11	11	1	0	0	1

Figure B.4: **Combinational circuit represented a truth table.** The truth table of the combinational circuit performing 2-bit multiplication.

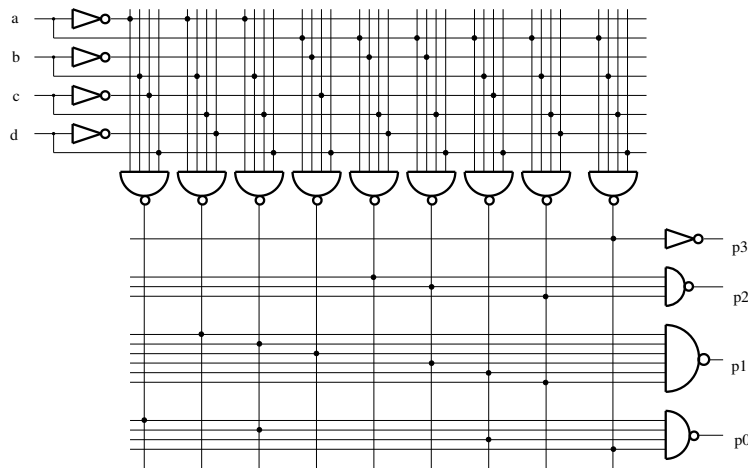


Figure B.5: **Direct implementation of a combinational circuit.** The direct implementation starting from DNF of the 2-bit multiplier.

B.4.2 Algebraic minimization

Minimal depth minimization

Example B.2 *Let's revisit the previous example for minimizing independently each function. The least significant output has the following form:*

$$p_0 = a'bc'd + a'bcd + abc'd + abcd.$$

We will apply the following steps:

$$p_0 = (a'bd)c' + (a'bd)c + (abd)c' + (abd)c$$

to emphasize the possibility of applying twice the exclusion principle, resulting

$$p_0 = a'bd + abd.$$

Applying again the same principle results:

$$p0 = bd(a' + a) = bd1 = bd.$$

The exclusion principle allowed us to reduce the size of the circuit from 22 to 2.

We continue with the next output:

$$p1 = a'bcd' + a'bcd + ab'c'd + ab'cd + abc'd + abcd' =$$

$$= a'bc(d' + d) + ab'd(c' + c) + abc'd + abcd' =$$

$$= a'bc + ab'd + abc'd + abcd' =$$

$$= bc(a' + ad') + ad(b' + bc') =$$

$$= bc(a' + d') + ad(b' + c') =$$

$$= a'bc + bcd' + ab'd + ac'd.$$

Now we used also the half-absorption principle reducing the size from 28 to 16.

Follows the minimization of $p2$:

$$p2 = ab'cd' + ab'cd + abcd' =$$

$$= ab'c + abcd' =$$

$$= ab'c + acd'$$

The $p3$ output can not be minimized. De Morgan law is used to transform the expressions to be implemented with NANDs.

$$p3 = ((abcd)')$$

$$p2 = ((ab'c)'(acd')')$$

$$p1 = ((a'bc)'(bcd')'(ab'd)'(ac'd)')$$

$$p0 = ((abcd)')$$

Results the circuit from Figure B.6. \diamond

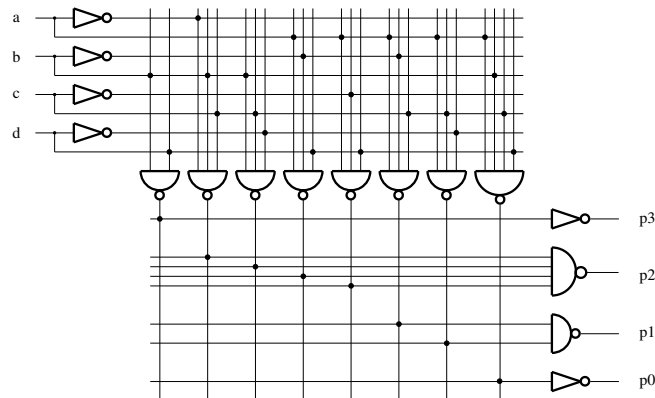


Figure B.6: **Minimal depth minimization** The first, minimal depth minimization of the 2-bit multiplier.

Multi-level minimization

Example B.3 *The same circuit for multiplying 2-bit numbers is used to exemplify the multilevel minimization. Results:*

$$p3 = abcd$$

$$p2 = ab'c + acd' = ac(b' + d') = ac(bd)'$$

$$p1 = a'bc + bcd' + ab'd + ac'd = bc(a' + d') + ad(b' + c') = bc(ad)' + ad(bc)' = (bc) \oplus (ad)$$

$$p0 = bd.$$

Using for XOR the following form:

$$x \oplus y = ((x \oplus y)')' = (xy + x'y) = (xy)'(x'y) = (xy)'(x + y)$$

results the circuit from Figure B.7 with size 22. \diamond

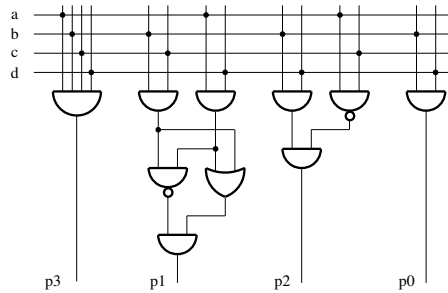


Figure B.7: **Multi-level minimization.** The second, multi-level minimization of the 2-bit multiplier.

Many output circuit minimization

Example B.4 *Inspecting carefully the schematics from Figure B.7 results: (1) the output p3 can be obtained inverting the NAND's output from the circuit of p2, (2) the output p0 is computed by a part of the circuit used for p2. Thus, we are encouraged to rewrite some of the functions in order to maximize the common circuits used in implementation. Results:*

$$x \oplus y = (xy)'(x + y) = ((xy) + (x + y)')'$$

$$p2 = ac(bd)' = ((ac)' + bd)'$$

allowing the simplified circuit from Figure B.8. The size is 16 and the depth is 3. But, more important: (1) the circuits contains only 2-input gates and (2) the maximum fan-out is 2. Both last characteristics led to small area and high speed. \diamond

B.4.3 Veitch-Karnaugh diagrams

In order to apply efficiently the exclusion principle we need to group carefully the minterms. Two dimension diagrams allow to emphasize the best grouping. Formally, the two minterms are adjacent if the Hamming distance is minimal.

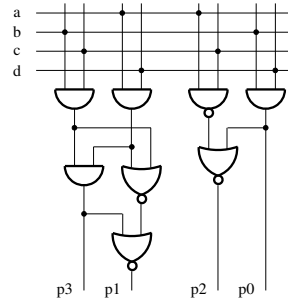


Figure B.8: **Multiple-output minimization.** The third, multiple-output minimization of the 2-bit multiplier.

Definition B.11 The Hamming distance between two minterms is given by the total numbers of binary variable which occur distinct in the two minterms. \diamond

Example B.5 The Hamming distance between $m_9 = ab'c'd$ and $m_4 = a'bc'd'$ is 3, because only the variable b occurs in the same form in both minterms.

The Hamming distance between $m_9 = ab'c'd$ and $m_1 = a'b'c'd$ is 1, because only the variable which occurs distinct in the two minterms is a . \diamond

Two n -variable terms having the Hamming distance 1 are minimized, using the exclusion principle, to one $(n - 1)$ -variable term. The size of the associated circuit is reduced from $2(n + 1)$ to $n - 1$.

A n -input Veitch diagram is a two dimensional surface containing 2^n squares, one for each n -value minterm. The adjacent minterms (minterms having the Hamming distance equal with 1) are placed in adjacent squares. In Figure B.9 are presented the Veitch diagrams for 2, 3 and 4-variable logic functions. For example, the 4-input diagram contains in the left half all minterms true for $a = 1$, in the upper half all minterms true for $b = 1$, in the two middle columns all the minterms true for $c = 1$, and in the two middle lines all the minterms true for $d = 1$. Results the lateral columns are adjacent and the lateral line are also adjacent. Actually the surface can be seen as a toroid.

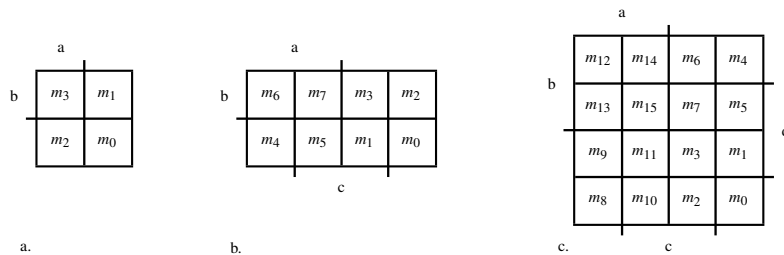


Figure B.9: **Veitch diagrams.** The Veitch diagrams for 2, 3, and 4 variables.

Example B.6 Let be the function $p1$ and $p2$, two outputs of the 2-bit multiplier. Rewriting them using minterms results::

$$p1 = m_6 + m_7 + m_9 + m_{11} + m_{13} + m_{14}$$

$$p2 = m_{10} + m_{11} + m_{14}.$$

In Figure B.10 $p1$ and $p2$ are represented.

◇

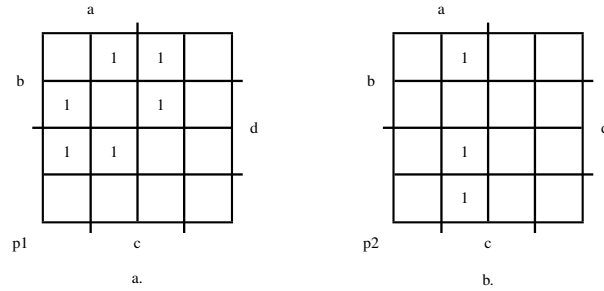


Figure B.10: **Using Veitch diagrams.** The Veitch diagrams for the functions $p1$ and $p2$.

The Karnaugh diagrams have the same property. The only difference is the way in which the minterms are assigned to squares. For example, in a 4-input Karnaugh diagram each column is associated to a pair of input variable and each line is associated with a pair containing the other variables. The columns are numbered in Gray sequence (successive binary configurations are adjacent). The first column contains all minterms true for $ab = 00$, the second column contains all minterms true for $ab = 01$, the third column contains all minterms true for $ab = 11$, the last column contains all minterms true for $ab = 10$. A similar association is made for lines. The Gray numbering provides a similar adjacency as in Veitch diagrams.

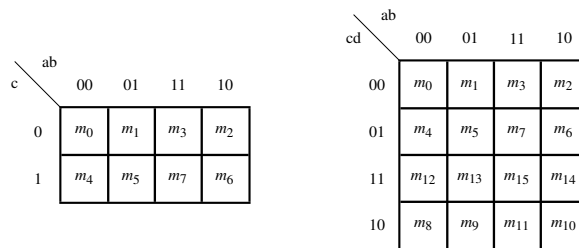


Figure B.11: **Karnaugh diagrams.** The Karnaugh diagrams for 3 and 4 variables.

In Figure B.12 the same functions, $p1$ and $p2$, are represented. The distribution of the surface is different but the degree of adjacency is identical.

In the following we will use Veitch diagrams, but we will name the them **V-K diagrams** to be fair with both Veitch and Karnaugh.

Minimizing with V-K diagrams

The rule to extract the minimized form of a function from a V-K diagram supposes:

- to define:
 - the *smallest* number

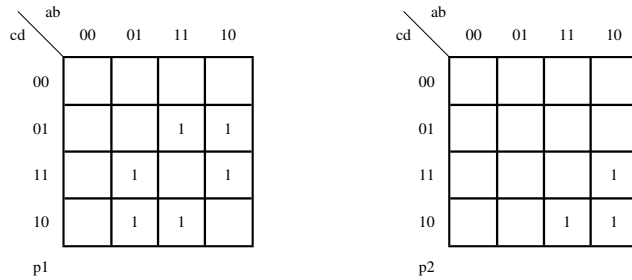


Figure B.12: Using Karnaugh diagrams. The Karnaugh diagrams for the functions p1 and p2.

- of rectangular surfaces containing only 1's
 - including all the 1's
 - each surface having a maximal area
 - and containing a power of two number of 1's
- to extract the logic terms (logic product of Boolean variables) associated with each previously emphasized surface
 - to provide the minimized function adding logically (logical OR function) the terms associated with the surfaces.

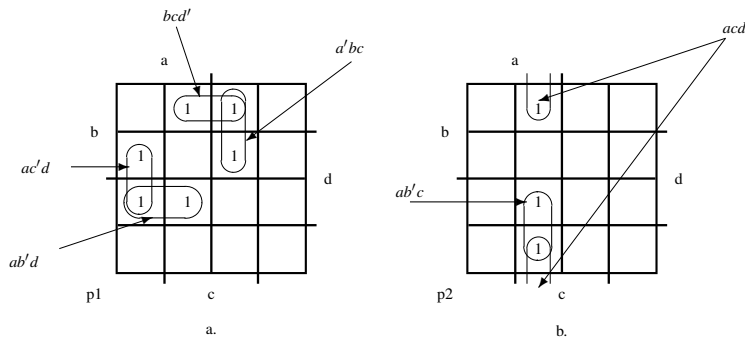


Figure B.13: Minimizing with V-K diagrams. Minimizing the functions p1 and p2.

Example B.7 Let's take the V-K diagrams from Figure B.10. In the V-K diagram for p1 there are four 2-square surfaces. The upper horizontal surface is included in the upper half of V-K diagram where $b = 1$, it is also included in the two middle columns where $c = 1$ and it is included in the surface formed by the two horizontal edges of the diagram where $d = 0$. Therefore, the associated term is bcd' which is true for: $(b = 1)AND(c = 1)AND(d = 0)$.

Because the horizontal edges are considered adjacent, in the V-K diagram for p2 m_{14} and m_{10} are adjacent forming a surface having acd' as associated term.

The previously known form of p1 and p2 result if the terms resulting from the two diagrams are logically added. \diamond

Minimizing incomplete defined functions

There are logic functions incompletely defined, which means for some binary input configurations the output value does not matter. For example, the designer knows that some inputs do not occur anytime. This lack in definition can be used to make an advanced minimization. In the V-K diagrams the corresponding minterms are marked as “don’t care”’s with “-”. When the surfaces are maximized the “don’t care”’s can be used to increase the area of 1’s. Thus, some “don’t care”’s will take the value 1 (those which are included in the surfaces of 1’s) and some of “don’t care”’s will take the value 0 (those which are not included in the surfaces of 1’s).

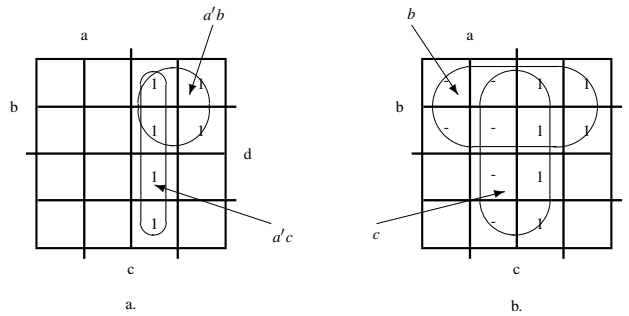


Figure B.14: **Minimizing incomplete defined functions.** **a.** The minimization of y (Example 1.8) ignoring the “don’t care” terms. **b.** The minimization of y (Example 1.8) considering the “don’t care” terms.

Example B.8 Let be the 4-input circuit receiving the binary coded decimals (from 0000 to 1001) indicating on its output if the received number is contained in the interval $[2, 7]$. It is supposed the binary configurations from 1010 to 1111 are not applied on the input of the circuit. If by hazard the circuit receives a meaningless input we do not care about the value generated by the circuit on its output.

In Figure B.14a the V-K diagram is presented for the version ignoring the “don’t care”’s. Results the function: $y = a'b + a'c = a'(b + c)$.

If “don’t care”’s are considered results the V-K diagram from Figure B.14b. Now each of the two surfaces are doubled resulting a more simplified form: $y = b + c$. \diamond

V-K diagrams with included functions

For various reasons in a V-K diagram we need to include instead of a logic value, 0 or 1, a logic function of variables which are different from the variables associated with the V-K diagram. For example, a minterm depending on a, b, c, d can be defined as taking a value which is depending on another logic 2-variable function by s, t .

A *simplified rule* to extract the minimized form of a function from a V-K diagram containing included functions is the following:

1. consider first only the 1s from the diagram and the rest of the diagram filled only with 0s and extract the resulting function
2. consider the 1s as “don’t care”’s for surfaces containing the same function and extract the resulting function “multiplying” the terms with the function

3. “add” the two functions.

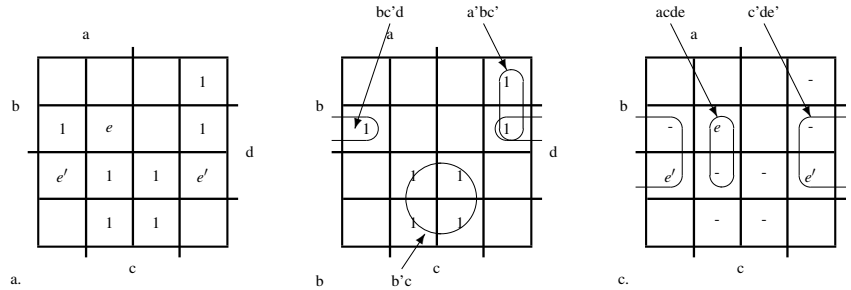


Figure B.15: An example of V-K diagram with included functions. **a.** The initial form. **b.** The form considered in the first step. **c.** The form considered in the second step.

Example B.9 Let be the function defined in Figure B.15a. The first step means to define the surfaces of 1s ignoring the squares containing functions. In Figure B.15b are defined 3 surfaces which provide the first form depending only by the variables a, b, c, d :

$$bc'd + a'bc' + b'c$$

The second step is based on the diagram represented in Figure B.15c, where a surface ($c'd$) is defined for the function e' and a smaller one ($acde$) for the function e . Results:

$$c'de' + acde$$

In the third step the two forms are “added” resulting:

$$f(a, b, c, d, e) = bc'd + a'bc' + b'c + c'de' + acde.$$

◇

Sometimes, an additional algebraic minimization is needed. But, it deserves because including functions in V-K diagrams is a way to expand the number of variable of the functions represented with a manageable V-K diagram.

B.5 Problems

Problem B.1

Appendix C

Introduction in ADC & DAC Convertors

This appendix contains a brief introduction to AD conversion and DA conversion. The aim is to give a preliminary picture of what it means to convert from analog to digital and vice versa. Presentation involves knowledge of the concept of operational amplifier and how it is used to deal with a comparator and a voltage amplifier. Also, the function of the digital priority encoder circuit must be known (see subsection 2.1.4).

C.1 Analog circuits

The operational amplifier is a concept that refers to an ideal circuit that is quite well approximated by real circuits.

Figure C.1 shows the symbol used for the operational amplifier. In the ideal case the amplification A is infinite (in reality it is very large, usually 10,000+). Another important characteristic of operational amplifiers is that they have a high input impedance Z_{in} . Input impedance is measured between the negative and positive input terminals, and its ideal value is infinity, which minimizes loading of the source. Also, an operational amplifier ideally has zero output impedance, Z_{out} .

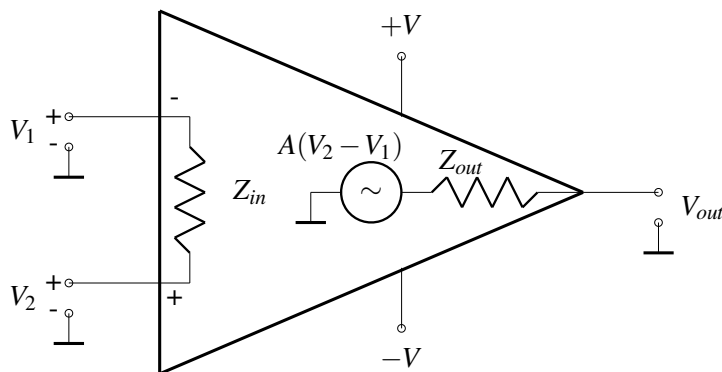


Figure C.1: Operational amplifier

We will use the operational amplifier in two established configurations: to implement the analog

comparison function and to perform the amplification used for the analog summation.

The operation of an analog comparator (see Figure C.2a) is the generation of binary-valued voltages that switch between the two levels when an analog input crosses a threshold voltage, V_{th} . Because

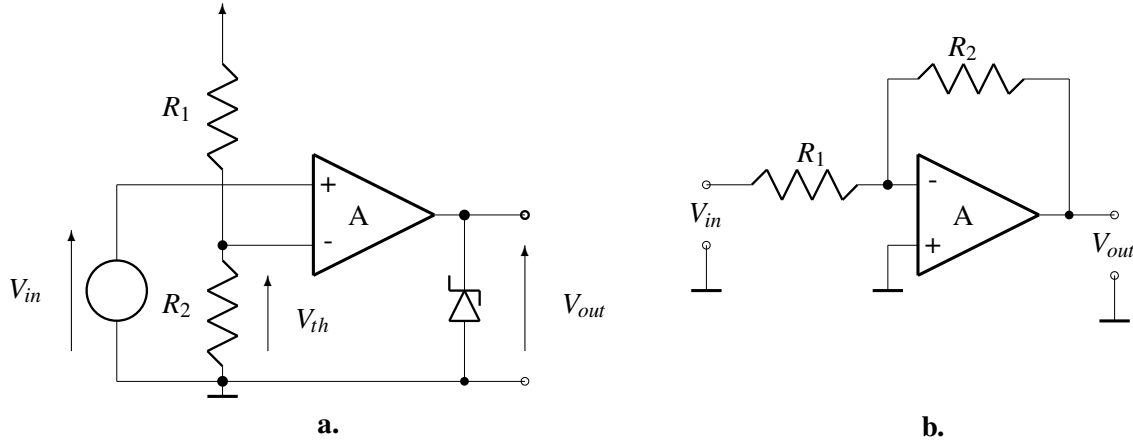


Figure C.2: Operational amplifier applications. **a.** Analog comparator. **b.** Amplifier.

$$V_{out} = A(V_{in} - V_{th})$$

a practical approximate model for the comparator is given by:

$$\begin{aligned} V_{out} &= V_z \text{ for } V_{in} > V_{th} \\ V_{out} &\simeq 0 \text{ for } V_{in} < V_{th} \end{aligned}$$

where V_z is the Zener voltage. Because A is infinite (actually very big) the output switches as soon as the input value reaches the threshold value, ensuring a very accurate threshold detection.

An inverting operational amplifiers (see Figure C.2b) is based on the fact that the operational amplifiers forces the negative terminal to equal the positive terminal, which is connected to ground. Indeed, the very high value of A generates an appropriate value on the output for a very small, practically zero, value of $V_1 - V_2$. Thus, V_2 , the inverting input, is practically connected to zero. Therefore the currents flowing through the resistors R_1 and R_2 are identical. Results:

$$\frac{V_{in}}{R_1} = -\frac{V_{out}}{R_2}$$

and the transfer function of the inverting amplifier is:

$$a = \frac{V_{out}}{V_{in}} = -\frac{R_2}{R_1}$$

C.2 ADC

The analog-digital conversion is based on the use of comparators and a resistor network. The accuracy with which the conversion is performed depends on the accuracy with which the resistance of the resistors is ensured and on the accuracy with which the comparators work.

For $V_{in} = 0$ all comparators have zero output. For $V_{in} > 0$ a number of comparators are activated and the encoder inputs are active from I_0 to I_i . Then the output of the encoder will generate the number i represented in binary code.

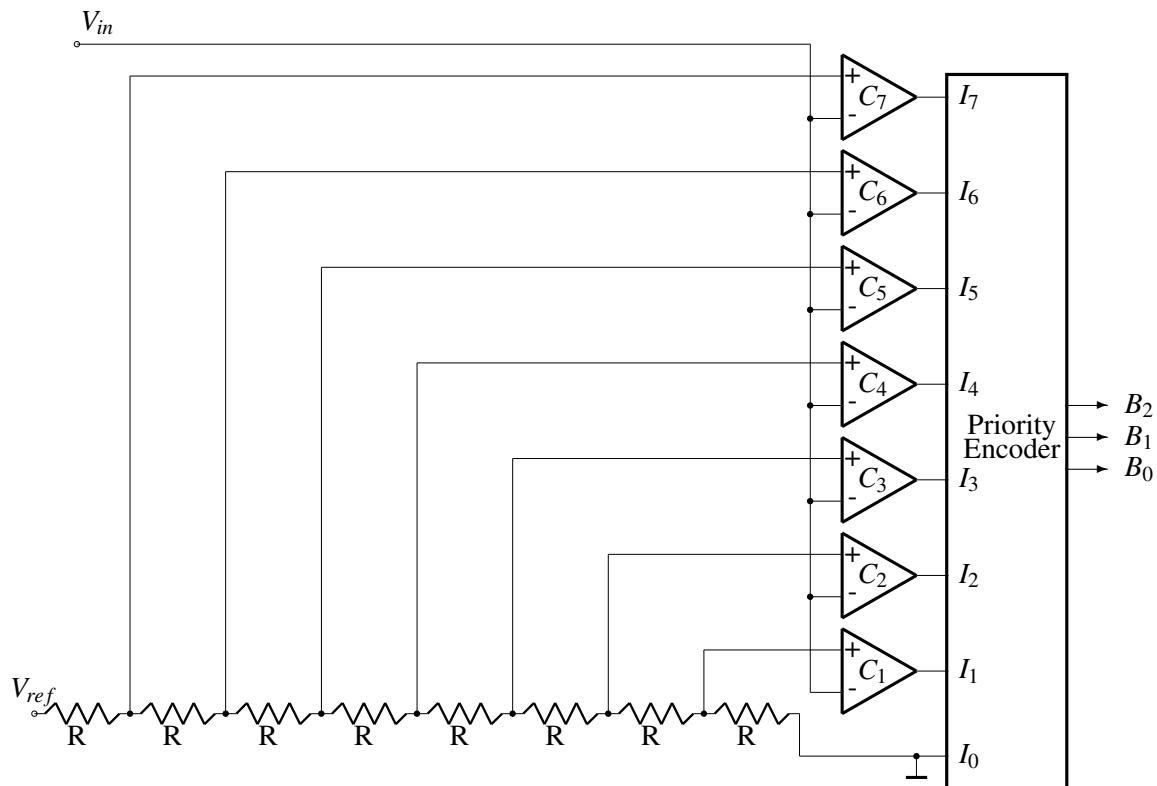


Figure C.3: ADC

C.3 DAC

For digital-to-analog conversion, a multi-input amplifier is used that allows the summation of several currents passing through resistors subjected to the same potential. The size of the resistors is inversely proportional to the associated binary order. Figure C.4 shows a DAC that converts 3-bit binary numbers. MSB is associated with the lowest resistance, of R value. The middle bit controls the current through a $2R$ value resistor, and the LSB commands a $4R$ value resistor. The sum of the currents passing through these resistors is equal to the current flowing through the reaction resistor R connected from the output of the operational amplifier to its reversing input.

If B_i , for $i = 0, 1, 2$, takes value in the set $\{0, 1\}$ and the truth value 0 is represented by 0 V and the truth

value 1 is represented by V_{DD} , then because the input current on the inverting input of the operational amplifier is zero we can write:

$$\frac{B_0}{2^2} + \frac{B_1}{2^1} + \frac{B_2}{2^0} = -\frac{V_{out}}{R}$$

and the output of the circuit represented in Figure C.4 results:

$$V_{out} = -V_{DD}(B_2/2^0 + B_1/2^1 + B_0/2^2)$$

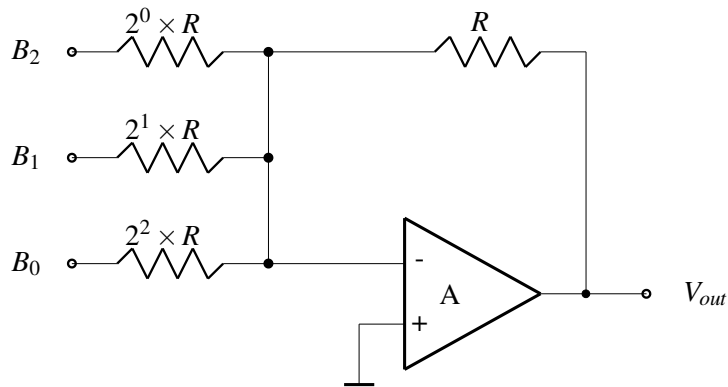


Figure C.4: DAC

For example, if $\{B_2, B_1, B_0\} = 101$, then the value on the output of the amplifier is: $1.25V_{DD}$.

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