

DIGITAL ELECTRONICS AND MICROPROCESSOR TH3

5th SEM ELECTRICAL ENGG.

PREPARED BY:

ER. BIKRAM KESHARI PARIDA

**[Lecturer Dept. of EE, KALINGA NAGAR POLYTECHNIC, TARAPUR,
JAJPUR ROAD]**

1 Basic Digital Concepts

By converting continuous analog signals into a finite number of discrete states, a process called *digitization*, then to the extent that the states are sufficiently well separated so that noise does not create errors, the resulting digital signals allow the following (slightly idealized):

- storage over arbitrary periods of time
- flawless retrieval and reproduction of the stored information
- flawless transmission of the information

Some information is intrinsically digital, so it is natural to process and manipulate it using purely digital techniques. Examples are numbers and words.

The drawback to digitization is that a single analog signal (*e.g.* a voltage which is a function of time, like a stereo signal) needs many discrete states, or *bits*, in order to give a satisfactory reproduction. For example, it requires a minimum of 10 bits to determine a voltage at any given time to an accuracy of $\approx 0.1\%$. For transmission, one now requires 10 lines instead of the one original analog line.

The explosion in digital techniques and technology has been made possible by the incredible increase in the density of digital circuitry, its robust performance, its relatively low cost, and its speed. The requirement of using many bits in reproduction is no longer an issue: The more the better.

This circuitry is based upon the transistor, which can be operated as a switch with two states. Hence, the digital information is intrinsically *binary*. So in practice, the terms digital and binary are used interchangeably. In the following sections we summarize some conventions for defining the binary states and for doing binary arithmetic.

Binary Logic States

The following table attempts to make correspondences between conventions for defining binary logic states. In the case of the TTL logic gates we will be using in the lab, the low voltage state is roughly 0–1 Volt and the high state is roughly 2.5–5 Volts. See page 475 of the text for the exact conventions for TTL as well as other hardware gate technologies.

Boolean Logic	Boolean Algebra	Voltage State (positive true)	Voltage State (negative true)
True (T)	1	High (H)	Low (L)
False (F)	0	L	H

The convention for naming these states is illustrated in Fig. 1. The “positive true” case is illustrated. The relationship between the logic state and label (in this case “switch open”) at some point in the circuit can be summarized with the following:

The labelled voltage is High (Low) when the label’s stated function is True (False). In the figure, the stated function is certainly true (switch open), and this does correspond to a high voltage at the labelled point. (Recall that with the switch open, Ohm’s Law implies that with zero current, the voltage difference across the “pullup” resistor is zero, so that

the labelled point is at +5 Volts. With a closed switch, the labelled point is connected to ground, with a 5 Volt drop across the resistor and a current of $I = V/R = 5 \text{ mA}$ through it.)

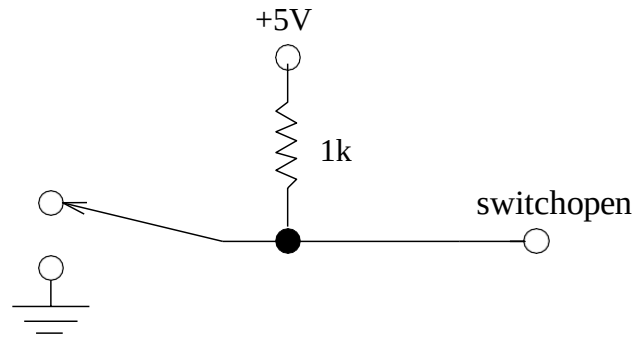


Figure 1: Illustration for labelling logic states (“positive true”).

With the convention known as “negative true”, the label would be changed to “switch closed” with a bar over it: switch closed. Our statement becomes:

The labelled voltage is Low (High) when the label’s stated function is True (False).

So in the figure, the stated function (switch closed) is true when the voltage is low. The bar is meant to evoke the boolean inversion operation: $\bar{T} = F$, $\bar{F} = T$, $\bar{\bar{T}} = T$, and so forth.

Binary Arithmetic

Each digit in binary is a 0 or a 1 and is called a *bit*, which is an abbreviation of *binary digit*. There are several common conventions for representation of numbers in binary.

The most familiar is *unsigned binary*. An example of a 8-bit number in this case is

$$01001111_2 = 0 \times 2^7 + 1 \times 2^6 + \dots + 1 \times 2^0 = 64 + 8 + 4 + 2 + 1 = 79_{10}$$

(Generally the subscripts will be omitted, since it will be clear from the context.) To convert from base 10 to binary, one can use a decomposition like above, or use the following algorithm illustrated by 79: $79/2 = 39$, remainder 1, then $39/2 = 19$ r 1, and so forth. Then assemble all the remainders in reverse order.

The largest number which can be represented by n bits is $2^n - 1$. For example, with 4 bits the largest number is $1111_2 = 15$.

The most significant bit (MSB) is the bit representing the highest power of 2, and the LSB represents the lowest power of 2.

Arithmetic with unsigned binary is analogous to decimal. For example 1-bit addition and multiplication are as follows: $0+0=0$, $0+1=1$, $1+1=0$, $0 \times 0=0$, $0 \times 1=0$, and $1 \times 1=1$. Note that this is *different from Boolean algebra*, as we shall see shortly, where $1+1 = 1$.

Another convention is called *BCD* (“binary coded decimal”). In this case each decimal digit is separately converted to binary. Therefore, since $7 = 0111_2$ and $9 = 1001_2$, then $79 = 01111001$ (BCD). Note that this is *different* than our previous result. We will use BCD quite often in this course. It is quite convenient, for example, when decimal numerical displays are used.

Yet another convention is *Gray code*. You have a homework problem to practice this. This is less commonly used.

Representation of Negative Numbers

There are two commonly used conventions for representing negative numbers.

With *sign magnitude*, the MSB is used to flag a negative number. So for example with 4-bit numbers we would have 0011=3 and 1011=−3. This is simple to see, but is not good for doing arithmetic.

With *2's complement*, negative numbers are designed so that the sum of a number and its 2's complement is zero. Using the 4-bit example again, we have 0101=5 and its 2's complement −5=1011. Adding (remember to carry) gives 10000=0. (The 5th bit doesn't count!) Both addition and multiplication work as you would expect using 2's complement. There are two methods for forming the 2's complement:

1. Make the transformation $0 \rightarrow 1$ and $1 \rightarrow 0$, then add 1.
2. Add some number to -2^{MSB} to get the number you want. For 4-bit numbers an example of finding the 2's complement of 5 is $-5 = -8 + 3 = 1000 + 0011 = 1011$.

Hexadecimal Representation

It is very often quite useful to represent blocks of 4 bits by a single digit. Thus in base 16 there is a convention for using one digit for the numbers 0, 1, 2, ..., 15 which is called *hexadecimal*. It follows decimal for 0–9, then uses letters A–F.

Decimal	Binary	Hex
0	0000	0
1	0001	1
2	0010	2
3	0011	3
4	0100	4
5	0101	5
6	0110	6
7	0111	7
8	1000	8
9	1001	9
10	1010	A
11	1011	B
12	1100	C
13	1101	D
14	1110	E
15	1111	F

2 Logic Gates and Combinational Logic

Gate Types and Truth Tables

The basic logic gates are AND, OR, NAND, NOR, XOR, INV, and BUF. The last two are not standard terms; they stand for “inverter” and “buffer”, respectively. The symbols for these gates and their corresponding Boolean expressions are given in Table 8.2 of the text which, for convenience, is reproduced (in part) in Fig. 2.

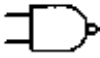
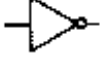
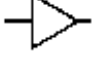
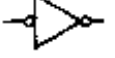
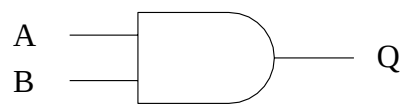
Name	Expression	Symbol	Negative true symbol
AND	AB		
NAND	$\overline{AB}$		
OR	$A + B$		
NOR	$\overline{A + B}$		
INVERT	$\bar{A}$		
BUFFER	A		
XOR	$A \oplus B$		
XNOR	$\overline{A \oplus B}$		

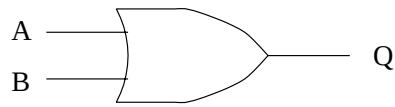
Figure 2: Table 8.2 from the text.

All of the logical gate functions, as well as the Boolean relations discussed in the next section, follow from the truth tables for the AND and OR gates. We reproduce these below. We also show the XOR truth table, because it comes up quite often, although, as we shall see, it is not elemental.



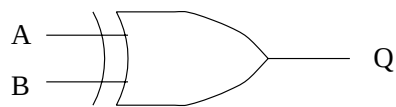
A	B	Q
0	0	0
1	0	0
0	1	0
1	1	1

Figure3:ANDgate.



A	B	Q
0	0	0
1	0	1
0	1	1
1	1	1

Figure4:ORgate.



A	B	Q
0	0	0
1	0	1
0	1	1
1	1	0

Figure5:XOR(exclusiveOR)gate.

Boolean Algebra and De Morgan's Theorems

Boolean algebra can be used to formalize the combinations of binary logic states. The fundamental relations are given in Table 8.3 of the text. In these relations, A and B are binary quantities, that is, they can be either logical true (T or 1) or logical false (F or 0). Most of these relations are obvious. Here are a few of them:

$$AA=A; \quad A+A=A; \quad A+A=1; \quad A\bar{A}=0; \quad \bar{\bar{A}}=A$$

Recall that the text sometimes uses an apostrophe for inversion (A'). We use the standard overbar notation ($\bar{A}$).

We can use algebraic expressions to complete our definition of the basic logic gates we began above. Note that the Boolean operations of "multiplication" and "addition" are defined by the truth tables for the AND and OR gates given above in Figs. 3 and 4. Using these definitions, we can define all of the logic gates algebraically. The truth tables can also be constructed from these relations, if necessary. See Fig. 2 for the gates symbols.

- AND: $Q=AB$ (see Fig. 3)
- OR: $Q=A+B$ (see Fig. 4)
- NAND: $Q=\overline{AB}$
- NOR: $Q=\overline{A+B}$
- XOR: $Q=A \oplus B$ (defined by truth table Fig. 5)
- INV: $Q=\bar{A}$
- BUF: $Q=A$

Example: Combining Gates

Let's re-express the XOR operation in terms of standard Boolean operations. The following truth table evaluates the expression $Q = \bar{A}B + A\bar{B}$.

A	B	$\bar{A}B$	$A\bar{B}$	Q
0	0	0	0	0
1	0	0	1	1
0	1	1	0	1
1	1	0	0	0

We see that this truth table is identical to the one for the XOR operation. Therefore, we can write

$$A \oplus B = \bar{A}B + A\bar{B} \quad (1)$$

A schematic of this expression in terms of gates is given in Fig. 6 (as well as Fig. 8.25 of the text). Recall that the open circles at the output or input of a gate represent inversion.

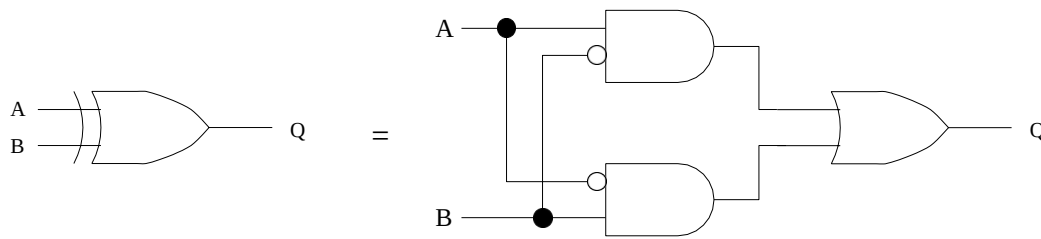


Figure 6: Realization of the XOR gate in terms of AND and OR gates.

Gate Interchangeability

In an example from the homework, we can make an INV gate from a 2-input NOR gate. Simply connect the two inputs of the NOR gate together. Algebraically, if the two original NOR gate inputs are labelled B and C , and they are recombined to form A , then we have $Q = B + C = A + A = \overline{A}$, which is the INV operation.

Note that an INV gate can not be made from OR or AND gates. For this reason the OR and AND gates are not universal. So for example, no combination of AND gates can be combined to substitute for a NOR gate. However, the NAND and NOR gates *are universal*.

DeMorgan

Perhaps the most interesting of the Boolean identities are the two known as DeMorgan's Theorems:

$$\overline{A+B} = \overline{A}\overline{B} \quad (\text{or, } A+B = \overline{\overline{A}\overline{B}}) \quad (2)$$

$$\overline{AB} = \overline{A} + \overline{B} \quad (\text{or, } AB = \overline{\overline{A} + \overline{B}}) \quad (3)$$

These expressions turn out to be quite useful, and we shall use them often.

An example of algebraic logic manipulation follows. It is the one mentioned at the end of Lab 1. One is to show that an XOR gate can be composed of 4 NAND gates. From the section above we know $A \oplus B = \overline{A}B + A\overline{B}$. Since $AA = A$ and $BB = B$, we can add these, rearrange, and apply the two DeMorgan relations to give

$$A \oplus B = A(A+B) + B(A+B) = A(AB) + B(AB) = A(AB)\overline{B} + B(AB)\overline{A}$$

Symbolic Logic

The two DeMorgan expressions above can be evoked using gate symbols by following this prescription: *Change gate shape (AND $\leftrightarrow$ OR) and invert all inputs and outputs.*

By examining the two rightmost columns of Fig. 2, one sees that the transformation between 3rd and 4th columns for the gates involving AND/OR gates works exactly in this way. For example, the DeMorgan expression $\overline{AB} = \overline{A} + \overline{B}$ is represented symbolically by the equivalence between the 3rd and 4th columns of the 2nd row ("NAND") of Fig. 2. We will go over how this works, and some more examples, in class.

Logic Minimization and Karnaugh Maps

As we found above, given a truth table, it is always possible to write down a correct logic expression simply by forming an OR of the ANDs of all input variables for which the output is true ($Q = 1$). However, for an arbitrary truth table such a procedure could produce a very lengthy and cumbersome expression which might be needlessly inefficient to implement with gates.

There are several methods for simplification of Boolean logic expressions. The process is usually called “logic minimization”, and the goal is to form a result which is efficient. Two methods we will discuss are algebraic minimization and Karnaugh maps. For very complicated problems the former method can be done using special software analysis programs. Karnaugh maps are also limited to problems with up to 4 binary inputs.

Let's start with a simple example. The table below gives an arbitrary truth table involving 2 logic inputs.

Table 1: Example of simple arbitrary truth table.

A	B	Q
0	0	1
0	1	1
1	0	0
1	1	1

There are two overall strategies:

1. Write down an expression directly from the truth table. Use Boolean algebra, if desired, to simplify.
2. Use Karnaugh mapping (“K-map”). This is only applicable if there are ≤ 4 inputs.

In our example above, we can use two different ways of writing down a result directly from the truth table. We can write down all TRUE terms and OR the result. This gives

$$Q = \bar{A}\bar{B} + \bar{A}B + AB$$

While correct, without further simplification this expression would involve 3 2-input AND gates, 2 inverters, and 1 3-input OR gate.

Alternatively, one can write down an expression for all of the FALSE states of the truth table. This is simpler in this case:

$$Q = A\bar{B} \rightarrow Q = A\bar{B} = \bar{A} + B$$

where the last step results from Eqn. 3. Presumably, the two expressions can be found to be equivalent with some algebra. Certainly, the 2nd is simpler, and involves only an inverter and one 2-input OR gate.

Finally, one can try a K-map solution. The first step is to write out the truth table in the form below, with the input states the headings of rows and columns of a table, and the corresponding outputs within, as shown below.

Table 2: K-map of truth table.

$A \setminus B$	0	1
0	1	1
1	0	1

The steps/rules are as follows:

1. Form the 2-dimensional table as above. Combine 2 inputs in a “graycode” way—see 2nd example below.
2. Form groups of 1's and circle them; the groups are rectangular and must have sides of length $2^n \times 2^m$, where n and m are integers $0, 1, 2, \dots$
3. The groups can overlap.
4. Write down an expression of the inputs for each group.
5. OR together these expressions. That's it.
6. Groups can wrap across table edges.
7. As before, one can alternatively form groups of 0's to give a solution for Q .
8. The bigger the group one can form, the better (simpler) the result.
9. There are usually many alternative solutions, all equivalent, some better than others depending upon what one is trying to optimize.

Here is one way of doing it:

$A \setminus B$	0	1
0	1	1
1	0	1

The two groups we have drawn are $\bar{A}$ and B . So the solution (as before) is:

$$Q = \bar{A} + B$$

2.4.1 K-map Example 2

Let's use this to determine which 3-bit numbers are prime. (This is a homework problem.)

We assume that 0, 1, 2 are *not* prime. We will let our input number have digits $a_2 a_1 a_0$. Here is the truth table:

Here is the corresponding K-map and a solution.

Note that where two inputs are combined in a row or column that their progression follows graycode, that is only one bit changes at a time. The solution shown above is:

$$Q = a_1 a_0 + a_2 a_0 = a_0(a_1 + a_2)$$

Table3:3-digitprimefinder.

Decimal	a_2	a_1	a_0	Q
0	0	0	0	0
1	0	0	1	0
2	0	1	0	0
3	0	1	1	1
4	1	0	0	0
5	1	0	1	1
6	1	1	0	0
7	1	1	1	1

Table4:K-mapoftruthtable.

$a_2 \backslash a_1 a_0$	00	01	11	10
0	0	0	1	0
1	0	1	1	0

K-map Example 3: Full Adder

In this example we will outline how to build a digital *full adder*. It is called “full” because it will include a “carry-in” bit and a “carry-out” bit. The carry bits will allow a succession of 1-bit full adders to be used to add binary numbers of arbitrary length. (A *half adder* includes only one carry bit.)

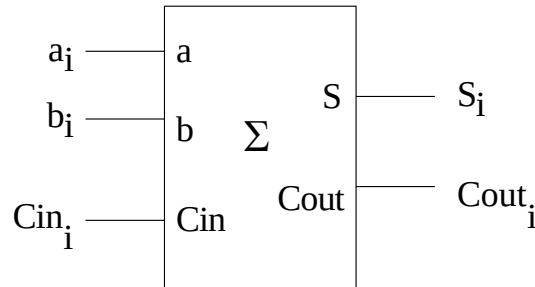


Figure 7: Block schematic of full adder. (We name our adder the “Σ chip”).

The scheme for the full adder is outlined in Fig. 7. Imagine that we are adding two n -bit binary numbers. Let the inputs a_i and b_i be the i -th bits of the two numbers. The carry in bit Cin_i represents any carry from the sum of the neighboring less significant bits at position $i-1$. That is, $Cin_i = 1$ if $a_{i-1} = b_{i-1} = 1$, and is 0 otherwise. The sum S_i at position i is therefore the sum of a_i , b_i , and Cin_i . (Note that this is an arithmetic sum, *not* a Boolean OR.) A carry for this sum sets the carry out bit, $Cout_i = 1$, which then can be applied to the sum of the $i+1$ bits. The truth table is given below.

Cin_i	a_i	b_i	S_i	$Cout_i$
0	0	0	0	0
0	0	1	1	0
0	1	0	1	0
0	1	1	0	1
1	0	0	1	0
1	0	1	0	1
1	1	0	0	1
1	1	1	1	1

With $Cin_i = 0$, we see that the output sum S_i is just given by the XOR operation, $a_i \oplus b_i$. And with $Cin_i = 1$, then $S_i = a_i \oplus b_i$. Perhaps the simplest way to express this relationship is the following:

$$S_i = Cin_i \oplus (a_i \oplus b_i)$$

To determine a relatively simple expression for $Cout_i$, we will use a K-map:

$Cin_i \backslash a_i b_i$	00	01	11	10
0	0	0	1	0
1	0	1	1	1

This yields

$$Cout_i = a_i b_i + Cin_i a_i + Cin_i b_i = a_i b_i + Cin_i (a_i + b_i)$$

which in hardware would be 2-input OR gates and 2-input AND gates.

As stated above, the carry bits allow our adder to be expanded to add any number of bits. As an example, a 4-bit adder circuit is depicted in Fig. 8. The sum can be 5 bits, where the MSB is formed by the final carryout. (Sometimes this is referred to as an “overflow” bit.)

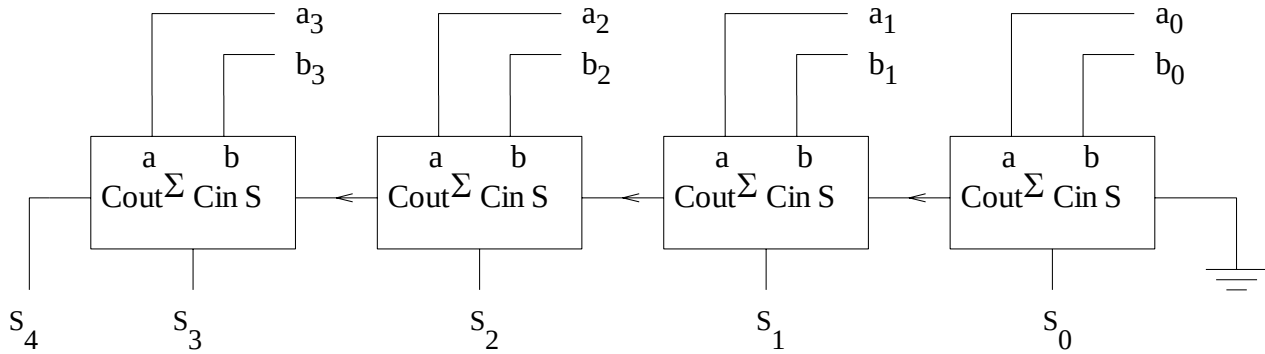


Figure 8: Expansion of 1-bit full adder to make a 4-bit adder.

Making a Multiplier from an Adder

In class we will discuss how to use our full adder (the “Σ chip”) to make a multiplier.

Multiplexing

A multiplexer (MUX) is a device which selects one of many inputs to a single output. The selection is done by using an input *address*. Hence, a MUX can take many data bits and put them, one at a time, on a single output data line in a particular sequence. This is an example of transforming *parallel* data to *serial* data. A demultiplexer (DEMUX) performs the inverse operation, taking one input and sending it to one of many possible outputs. Again the output line is selected using an address.

A MUX-DEMUX pair can be used to convert data to serial form for transmission, thus reducing the number of required transmission lines. The address bits are shared by the MUX and DEMUX at each end. If n data bits are to be transmitted, then after multiplexing, the number of separate lines required is $\log_2 n + 1$, compared to n without the conversion to serial. Hence for large n the saving can be substantial. In Lab 2, you will build such a system.

Multiplexers consist of two functionally separate components, a *decoder* and some switches or gates. The decoder interprets the input address to select a single data bit. We use the example of a 4-bit MUX in the following section to illustrate how this works.

A 4-bit MUX Design

We wish to design a 4-bit multiplexer. The block diagram is given in Fig. 9. There are 4 input data bits D_0 – D_3 , 2 input address bits A_0 and A_1 , one serial output data bit Q , and

an (optional) enable bit E which is used for expansion (discussed later). First we will design the decoder.

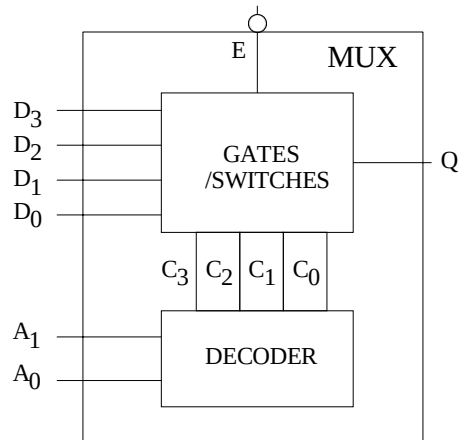


Figure9:Block diagram of 4-bit MUX.

We need m address bits to specify 2^m data bits. So in our example, we have 2 address bits. The truth table for our decoder is straightforward:

A_1	A_0	C_0	C_1	C_2	C_3
0	0	1	0	0	0
0	1	0	1	0	0
1	0	0	0	1	0
1	1	0	0	0	1

The implementation of the truth table with standard gates is also straightforward, as given in Fig.10.

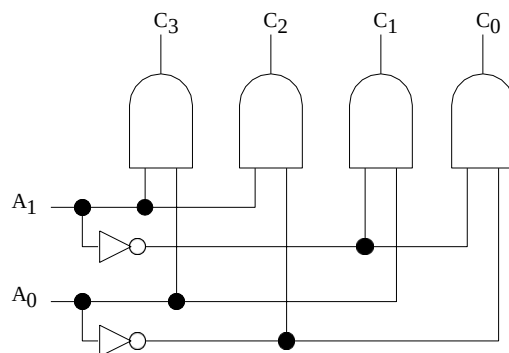


Figure10:Decoder for the 4-bit MUX.

For the “gates/switches” part of the MUX, the design depends upon whether the input data lines carry digital or analog signals. We will discuss the analog possibility later. The digital case is the usual and simplest case. Here, the data routing can be accomplished

simply by forming 2-input ANDs of the decoder outputs with the corresponding data input, and then forming an OR of these terms. Explicitly,

$$Q = C_0 D_0 + C_1 D_1 + C_2 D_2 + C_3 D_3$$

Finally, if an ENABLE line E is included, it is simply ANDed with the right hand side of this expression. This can be used to switch the entire MUXI off/on, and is useful for expansion to more bits as we shall see.

3 Flip-Flops and Introductory Sequential Logic

We now turn to digital circuits which have states which change in time, usually according to an external clock. The *flip-flop* is an important element of such circuits. It has the interesting property of *memory*: It can be set to a state which is retained until explicitly reset.

Simple Latches

The following 3 figures are equivalent representations of a simple circuit. In general these are called flip-flops. Specifically, these examples are called SR ("set-reset") flip-flops, or SR latches.

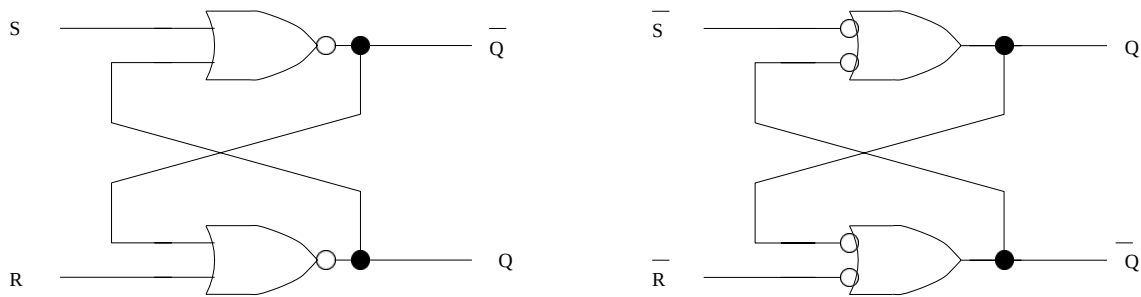


Figure 11: Two equivalent versions of an SR flip-flop (or "SR latch").

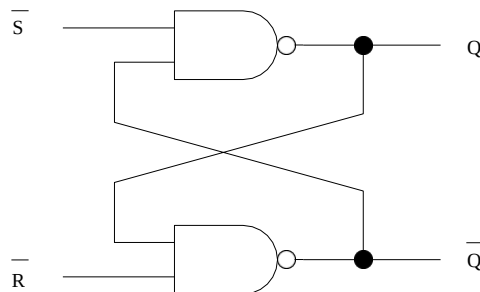


Figure 12: Yet another equivalent SR flip-flop, as used in Lab 3.

The truth table for the SR latch is given below.

S	$\bar{S}$	R	$\bar{R}$	Q	$\bar{Q}$
1	0	0	1	1	0
0	1	1	0	0	1
0	1	0	1	retains previous	
1	0	1	0	0	0

The state described by the last row is clearly problematic, since Q and $\bar{Q}$ should not be the same value. Thus, the $S = R = 1$ inputs should be avoided.

From the truth table, we can develop sequences such as the following:

1. $R=0, S=1 \Rightarrow Q=1$ (set)
2. $R=0, S=0 \Rightarrow Q=1$ ($Q=1$ state retained: “memory”)
3. $R=1, S=0 \Rightarrow Q=0$ (reset)
4. $R=0, S=0 \Rightarrow Q=0$ ($Q=0$ state retained)

In alternative language, the first operation “writes” a true state into one bit of memory. It can subsequently be “read” until it is erased by the reset operation of the third line.

Latch Example: Debounced Switch

A useful example of the simple SR flip-flop is the debounced switch, like the ones on the lab prototyping boards. The point is that any simple mechanical switch will bounce as it makes contact. Hence, an attempt to provide a simple transition from digital HIGH to LOW with a mechanical switch may result in an unintended series of transitions between the two states as the switch damps to its final position. So, for example, a digital counter connected to Q would count every bounce, rather than the single push of the button which was intended.

The debounced configuration and corresponding truth table are given below. When the switch is moved from A to B , for example, the output Q goes LOW. A bounce would result in $A = B = 1$, which is the “retain previous” state of the flip-flop. Hence, the bounces do not appear at the output Q .

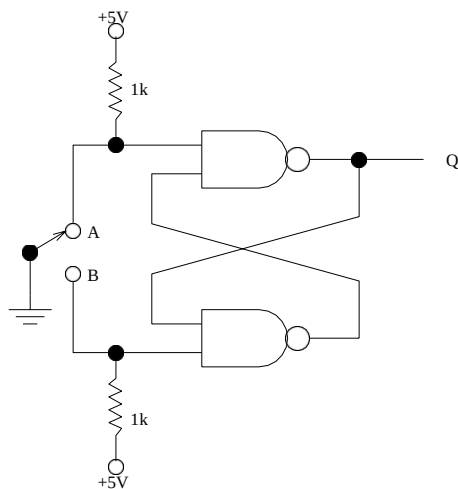


Figure 13: A debounced switch.

A	B	Q
1	0	0
0	1	1
1	1	retains previous
0	0	not allowed

Clocked Flip-flops

We will soon get used to the idea of a clock as an essential element of digital circuitry. When we speak of a clock signal, we mean a sequence of evenly spaced digital high and low signals proceeding at a fixed frequency. That is, the clock is a continuous sequence of square wave pulses. There are a number of reasons for the importance of the clock. Clearly it is essential for doing any kind of counting or timing operation. But, its most important role is in providing *synchronization* to the digital circuit. Each clock pulse may represent the transition to a new digital state of a so-called “state machine” (simple processor) we will soon encounter. Or a clock pulse may correspond to the movement of a bit of data from one location in memory to another. A digital circuit coordinates these various functions by the synchronization provided by a single clock signal which is shared throughout the circuit. A more sophisticated example of this concept is the clock of a computer, which we have come to associate with processing speed (e.g. 330 MHz for typical current generation commercial processors.)

We can include a clock signal to our simple SR flip-flop, as shown in Fig. 14. The truth table, given below, follows directly from our previous SR flip-flop, except now we include a label for the n^{th} clock pulse for the inputs and the output. This is because the inputs have no effect unless they coincide with a clock pulse. (Note that a specified clock pulse conventionally refers to a HIGH level.) As indicated in the truth table, the inputs $S_n = R_n = 0$ represent the flip-flop memory state. Significantly, one notes that the interval *between* clock pulses also corresponds to the “retain previous state” of the flip-flop. Hence the information encoded by the one bit of flip-flop memory can only be modified in synchronization with the clock.

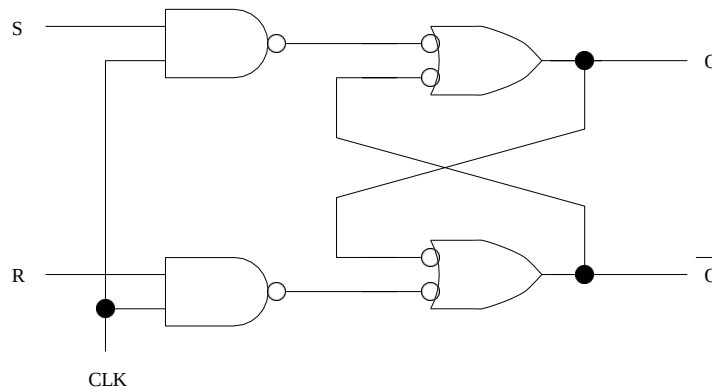


Figure 14: A clocked SR flip-flop.

S_n	R_n	Q_n
1	0	1
0	1	0
0	0	Q_{n-1}
1	1	avoid

We are now set to make a subtle transition for our next version of the clocked flip-flop. The flip-flop memory is being used to retain the state between clock pulses. In fact, the states set up by the S and R inputs can be represented by a single input we call “data”, or

D. This is shown in Fig.15. Note that we have explicitly eliminated the bad $S=R=1$ state with this configuration.

We can override this data input and clock synchronization scheme by including the “jam set” ($\bar{S}$) and “jam reset” ($\bar{R}$) inputs shown in Fig.15. These function just as before with the unclocked SR flip-flop. Note that these “jam” inputs go by various names. So sometimes the set is called “preset” and reset is called “clear”, for example.

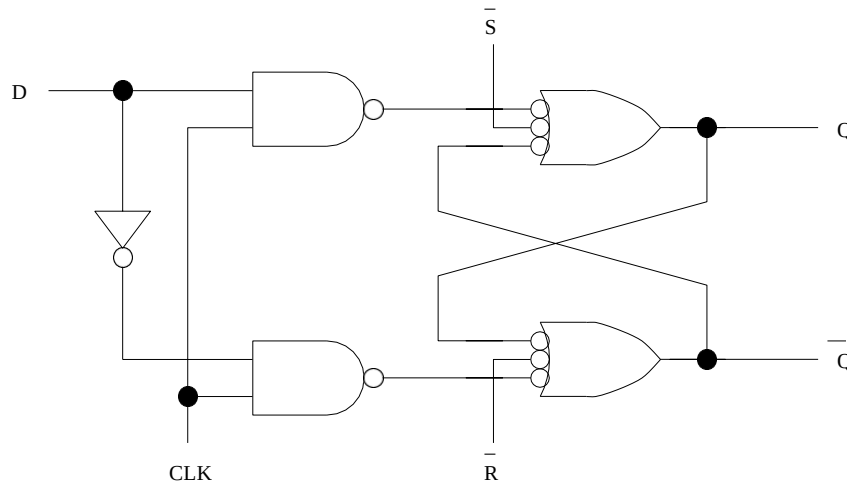


Figure15:A “D-type transparent” flip-flop with jam set and reset.

A typical timing diagram for this flip-flop is given in Fig.16. Note that the jam reset signal $\bar{R}$ overrides any action of the data or clock inputs.

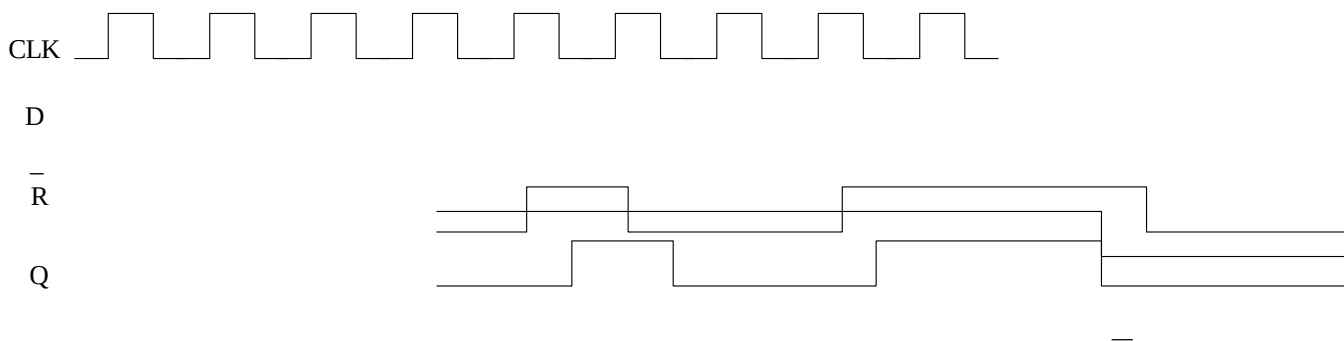


Figure16: Example of timing diagram for the transparent D flip-flop. (It is assumed that S is held HIGH throughout.)

Edge Triggered Flip-Flops

We need to make one final modification to our clocked flip-flop. Note that in the timing diagram of Fig.16 that there is quite a bit of apparent ambiguity regarding exactly when the D input gets latched into Q. If a transition in D occurs sometime during a clock HIGH, for example, what will occur? The answer will depend upon the characteristics of the particular electronics being used. This lack of clarity is often unacceptable. As a point of terminology,

the clocked flip-flop of Fig.15 is called a *transparent D-type* flip-flop or latch.(An example in TTL is the 7475 IC.)

The solution to this is the *edge-triggered* flip-flop. We will discuss how this works for one example in class. It is also discussed some in the text. Triggering on a clock rising or falling edge is similar in all respects to what we have discussed, except that it requires 2–3 coupled SR-type flip-flops, rather than just one clocked SR flip-flop. The most common type is the *positive-edge triggered D-type* flip-flop. This latches the D input upon the clock transition from LOW to HIGH. An example of this in TTL is the 7474 IC. It is also common to employ a *negative-edge triggered D-type* flip-flop, which latches the D input upon the clock transition from HIGH to LOW.

The symbols used for these three D-type flip-flops are depicted in Fig.17. Note that the small triangle at the clock input depicts positive-edge triggering, and with an inversion symbol represents negative-edge triggered. The JK type of flip-flop is a slightly fancier version of the D-type which we will discuss briefly later. Not shown in the figure are the jam set and reset inputs, which are typically included in the flip-flop IC packages. In timing diagrams, the clocks for edge-triggered devices are indicated by arrows, as shown in Fig.18.

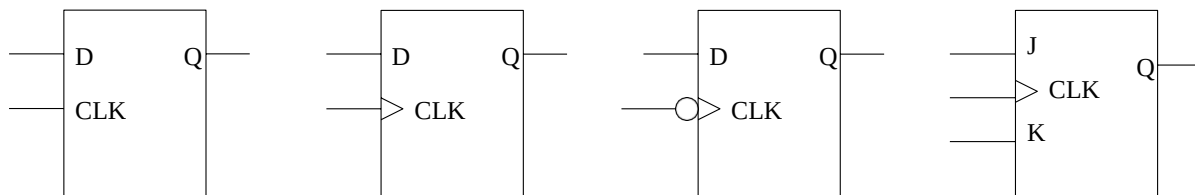


Figure 17: Symbols for D-type and JK flip-flops. Left to right: transparent D-type, positive-edge triggered D-type, negative-edge triggered D-type, and positive-edge triggered JK-type.



Figure 18: Clocks in timing diagrams for positive-edge triggered (left) and negative-edge triggered (right) devices.

For edge-triggered devices, the ambiguity regarding latch timing is reduced significantly. But at high clock frequency it will become an issue again. Typically, the requirements are as follows:

- The data input must be held for a time t_{setup} before the clock edge. Typically, $t_{\text{setup}} \approx 20\text{ ns}$ or less.
- For some ICs, the data must be held for a short time t_{hold} after the clock edge. Typically $t_{\text{hold}} \approx 3\text{ ns}$, but is zero for most newer ICs.
- The output Q appears after a short propagation delay t_{prop} of the signal through the gates of the IC. Typically, $t_{\text{prop}} \approx 10\text{ ns}$.

From these considerations we see that for clocks of frequency much less than $\sim 1/(10\text{ns}) = 100\text{ MHz}$, these issues will be unimportant, and we can effectively consider the transitions to occur instantaneously in our timing diagrams.

4 Counters, Registers, and State Machines

We can now apply what we know about basic flip-flops circuit elements to develop new functions: counters and registers. In doing so, we will introduce the “state machine”, a clocked sequential “processor”. We will examine this latter topic in more detail in a few weeks.

Divide by Two Counter

The edge-triggered D-type flip-flops which we introduced in the previous Section are quite useful and versatile building blocks of sequential logic. A simple application is the divide-by-2 counters shown in Fig. 19, along with the corresponding timing diagram.

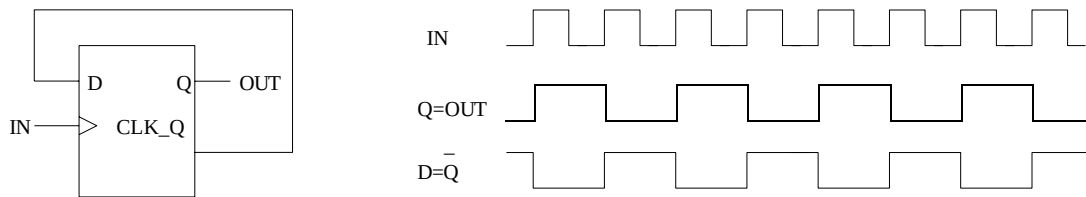


Figure 19: Positive edge-triggered D-type flip-flop connected as divide-by-2 counter.

Using the JK Flip-flop

In Lab 4 you will build an asynchronous (ripple) counter using a sequence of cascaded JK flip-flops, rather than the D-type which is used in our discussion below. For reference, the JK truth table is given in Fig. 20. Note that there is no fundamental advantage to using the JK instead of the D-type, only that the JK, with the additional $J=K=1$ state, makes the divide-by-2 function slightly simpler to implement.

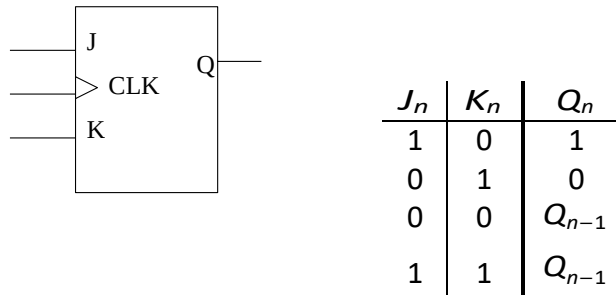


Figure 20: The JK Flip-flop.

AsynchronousCounter

Flip-flops can be connected in series, as shown in Fig.21. The resulting outputs are given in Fig.22. (Note that labels in these two figures correspond when $A \equiv 2^0$, $B \equiv 2^1$, $C \equiv 2^2$, and $D \equiv 2^3$. Hence, this is a 4-bit counter, with maximum count $2^4 - 1 = 15$. It is clearly possible to expand such a counter to an indefinite number of bits.

While asynchronous counters are easy to assemble, they have serious drawbacks for some applications. In particular, the input must propagate through the entire chain of flip-flops before the correct result is achieved. Eventually, at high input rate, the two ends of the chain, representing the LSB and MSB, can be processing different input pulses entirely. (Perhaps in lab you can see this effect on the oscilloscope with a very high input frequency.) The solution to this is the *synchronous counter*, which we will discuss below as an example of a state machine.

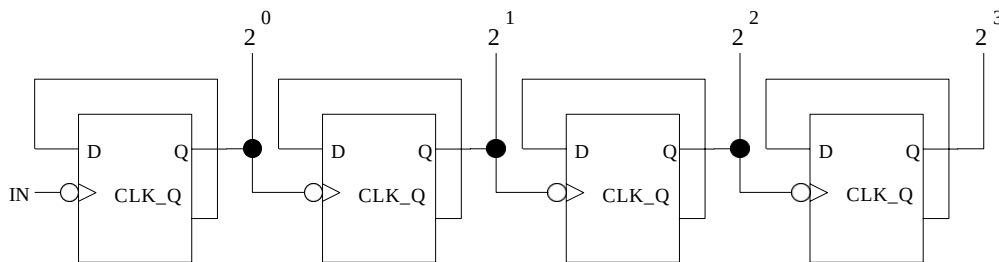


Figure 21: Asynchronous (“ripple”) counter made from cascaded D-type flip-flops.

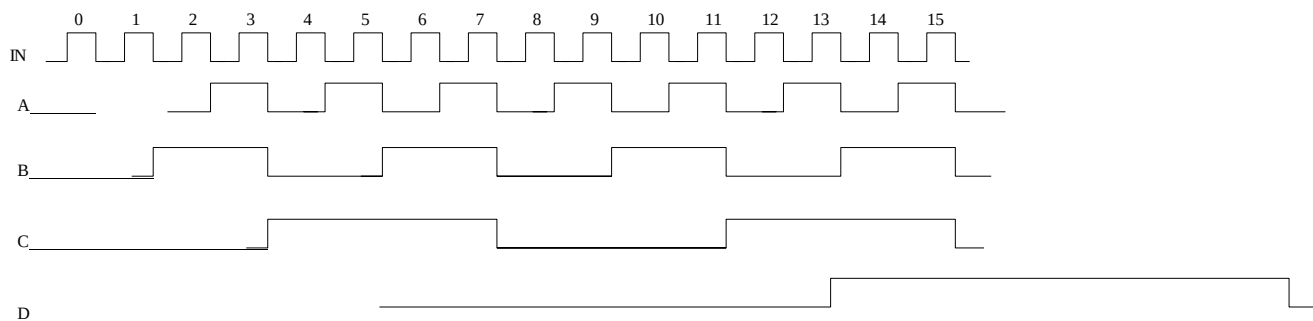


Figure 22: Waveforms generated by the ripple counter.

Registers

Basic Register

The figure below represents a 4-bit memory. We can think of it as 4 individual D-type flip-flops. The important point about a data register of this type is that all of the inputs are latched into memory synchronously by a single clock cycle.

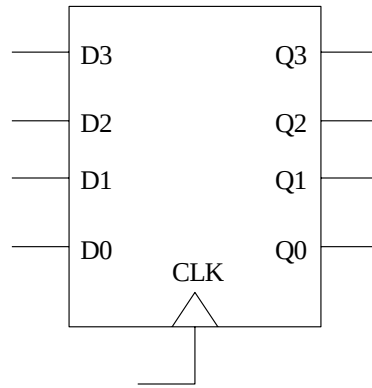


Figure 23: 4-bit data register.

Shift Registers

The figure below is an example of a 4-bit *shift register*. These configurations are quite useful, particularly for transforming serial data to parallel, and parallel to serial. In the circuit below, a pulse appearing at "serial in" would be shifted from the output of one flip-flop to the next one each clock cycle. Hence a serial bit pattern at the input (4 bits long in this example) would appear as 4 parallel bits in the outputs Q_0 – Q_3 after 4 clock cycles. This represents the serial-to-parallel case.

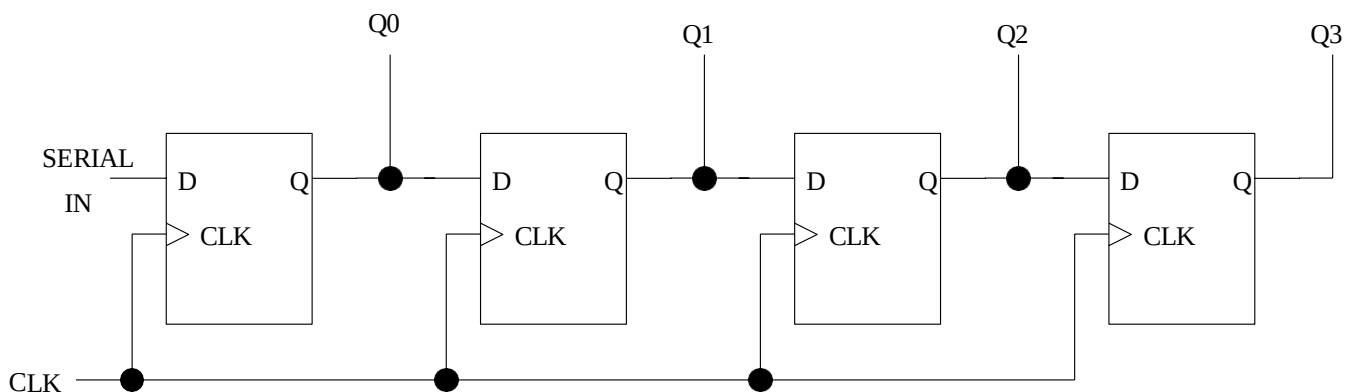


Figure 24: 4-bit shift register.

We will discuss several examples of shift registers a few lectures hence.

5 Analog/Digital Conversion

In this section we discuss the important topic of analog to digital conversion (often written A/D), and digital to analog conversion (D/A). On one hand, most electrical measurements are intrinsically analog. To take advantage of the great capabilities available for digital data storage, processing, and computation, on the other hand, requires the conversion of analog to digital. Hence, analog to digital (A/D) conversion techniques have become extremely important. A great deal of technical effort has gone into producing A/D converters (ADCs) which are fast, accurate, and cheap. D/A converters (DACs) are also very important. For example, video monitors convert digital information generated by computers to analog signals which are used to direct the electron beam at a specified portion of the monitor screen. DACs are conceptually simpler than ADCs, although it is difficult in practice to build a precise DAC. We will discuss D/A conversion before A/D. But first we go over some underlying ideas.

A/D Resolution

First of all we should keep in mind that there are several different schemes for encoding analog information as bits, depending upon what is required by a particular application. One extreme is that of encoding the complete analog signal in as much detail as possible. For example, a musical instrument produces an analog signal which is readily converted to an analog electrical signal using a microphone. If this is to be recorded digitally, one naturally would choose to digitize enough information so that when the recording is played back, the resulting audio is not perceived to be significantly different from the original. In this case the analog signal is a voltage which varies with time, $V(t)$.

At any time t_0 , $V(t_0)$ can be sampled and converted to digital. The analog signal must be sampled for a finite time, called the *sampling time*, Δt . One may guess that it is necessary to sample the analog signal continuously, with no gaps between consecutive samples. This turns out to be overkill. The *Nyquist Theorem* states that if the maximum frequency of interest in the analog input is $f_{\max}$, then perfect reproduction only requires that the sampling frequency f_{samp} be slightly greater than twice $f_{\max}$. That is,

$$f_{\text{samp}} > 2f_{\max}$$

For example, for audio signals the maximum frequency of interest is usually 20 kHz. In this case the input analog must be sampled at a little over 40 kHz. In fact, 44 kHz is typically used.

Alternatively, it might not be of interest to represent the entire analog input digitally. Perhaps only one feature of the analog signal is useful. One example is “peak sensing,” where one samples and digitizes the input only at the instant where an instrument’s output achieves a maximum analog output. Or one may average (“integrate”) an input signal over some predefined time, retaining only the average value to be digitized.

For any of these sampling schemes, there remains the issue of how many bits are to be used to describe the sampled signal $V(t_0)$. This is the question of *A/D resolution*. We need a standard definition of resolution. Let’s say, for example, that we choose to digitize the input using 12 bits. This means that we will try to match our analog input to 1 of $2^{12} = 4096$ possible levels. This is generally done by ascribing a number from 0 to 4095. So, assuming our ADC works correctly, the digital estimate of the analog input can, at worst, be wrong by the range of the LSB. On average, the error is half of this. This defines the resolution. Therefore, for our 12-bit example, the resolution is $1/(2 \cdot 4096)$, or a little worse than 0.01%.

D/A Conversion

The basic element of a DAC is the simplest analog divider: the resistor divider. First, we need to review the two important properties of an operational amplifier ("op-amp") connected in the inverting configuration. This is shown in Fig. 25. The two important properties are

1. The "−" input is effectively at ground. ("virtual ground")
2. The voltage gain is $G \equiv V_{out}/V_{in} = -R_2/R_1$. An equivalent statement is that for a current at the − input of $I_{in} = V_{in}/R_1$, the output voltage is $V_{out} = G V_{in} = -R_2 I_{in} = -V_{in} R_2/R_1$. Sometimes this is written in the form $V_{out} = g I_{in}$, where g is the *transconductance*, and $g = -R_2/R_1$ in this case.

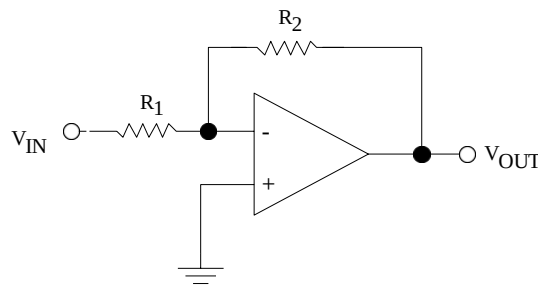


Figure 25: Inverting op-amp configuration.

The basic idea of most DACs is then made clear by the 4-bit example illustrated in Fig. 26. The input 4-bit digital signal defines the position of the switches labelled a_0 – a_3 . A HIGH input bit would correspond to a switch connected to 1.0V, whereas a LOW connects to ground. The configuration in the figure represents a binary input of 1010, or 10_{10} . Since the virtual ground keeps the op-amp input at ground, then for a switch connected to ground, there can be no current flow. However, for switches connected to 1.0 V, the current presented to the op-amp will be 1.0V divided by the resistance of that leg. All legs with HIGH switches then contribute some current. With the binary progression of resistance values shown in the figure, the desired result is obtained. So for the example shown, the total current to the op-amp is $I = 1.0/R + 1.0/(4R) = 5/(4R)$. The output voltage is

$$V_{out} = -RI = 5/4 = 1.25V$$

When all input bits are HIGH ($1111 = 15_{10}$), we find $V_{out} = 15/8V$. A simple check of our scheme shows that

$$(5/4)/(15/8) = 2/3 = 10/15 = 1010/1111$$

as expected.

The R-2R Ladder

This represents a rather minor point, although it is an interesting idea. The "R-2R ladder" is of practical interest because it uses only two resistor values. Since it is difficult to accurately fabricate resistors of arbitrary resistance, this is beneficial. The two resistances of the R-2R

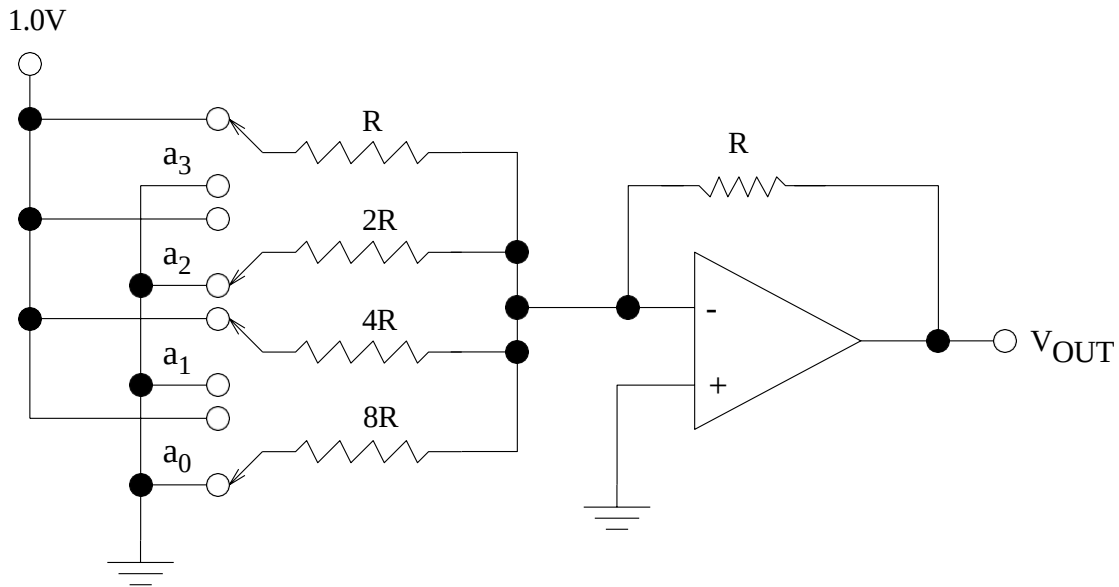


Figure 26: Example 4-bit DAC scheme.

are to be contrasted with the scheme represented by the circuit of Fig. 26, which employs as many resistance values as there are bits. The idea behind the R-2R ladder hinges on noticing the pattern of equivalences represented by Fig. 27, which can be used to replicate an arbitrarily long ladder, and hence handle an arbitrary number of bits.

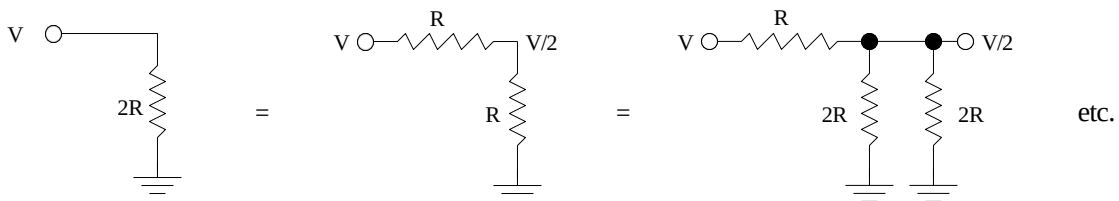


Figure 27: Principle of the R-2R ladder.
related with this equivalent circuit.

The rightmost 2R resistor can be indefinitely

A/D Conversion

ADCs fall into 3 general types of technique:

- (1) parallel encoding (flash): fast; limited accuracy
- (2) successive approx. (feedback): med. fast; good accuracy
- (3) single or double slope: slow; best potential accuracy

All of these techniques use a device known as a *comparator*. This was discussed in 431/531 and in the text Chapters 4 and 9. Here, we will not discuss how comparators work, but we do need to know what they do. There are many makes of comparators. We will use the model LM311 in lab. Figure 28 shows a comparator schematically. Internally, the comparator can be thought of as a fast, very high-gain differential amplifier ("A") with inputs "+" and "-." We can put a "threshold voltage" at the "-" input. Call it V_{th} . The circuit input V_{in} is connected to the "+" input. When $V_{in} > V_{th}$, the comparator amplifies this difference until the output reaches its largest possible value, which is determined by the connection through the pull-up resistor. In the configurations shown here, as well as in Lab 5, the $\sim 1k\Omega$ pull-up resistor is connected to +5 V. (Note that while +5 V is convenient for many digital circuits, it is possible to use other values, such as +12 V.) When $V_{in} < V_{th}$, the output swings the other way. This level is usually determined by a connection to one of the comparator pins. Here, it is ground.

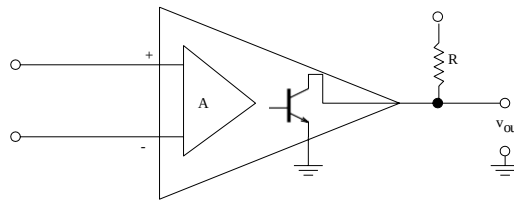


Figure 28: Comparator.

Hence, the comparator represents a one-bit ADC. When the analog input exceeds the pre-defined threshold, the output goes to digital HIGH, and when the input is less than the threshold, the output goes to digital LOW.

Flash ADCs

In this scheme, the input is fanned out in parallel to several comparators with monotonically increasing thresholds. The pattern of comparator outputs is then analyzed by some combinational logic (*i.e.* gates) to determine the output. This technique is called *flash* (or *parallel*) *encoding*. We exemplify the flash ADC scheme with the 2-bit ADC shown in Fig. 29. With $n=2$ bits, we need to define $2^n=4$ possible states. These states represent 4 separate intervals. The analog input will fall into one of these intervals, and we will encode this assignment with the 2 bits. Defining the boundaries of 2^n intervals requires 2^n-1 comparators, with the threshold of each comparator set to the appropriate boundary voltage.

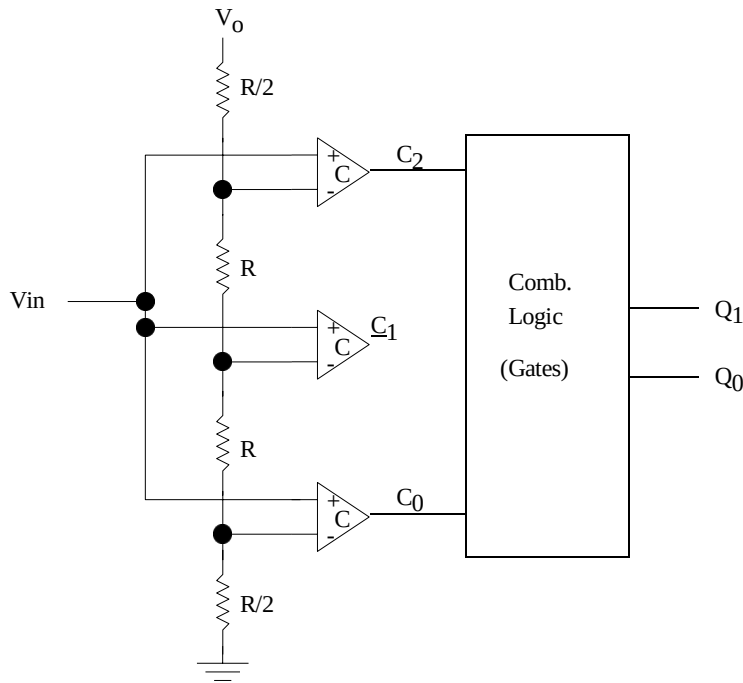


Figure29:Schematicofa2-bitflashADC.

Let's go through a concrete example. Assume that our FADC circuit is designed to handle analog voltage input signals in the range -0.5 to 3.5 V. Thus, we have a 4 -volt total input range, with each interval spanning 1.0 V. Therefore, each state will have a maximum error, or resolution, of half the interval, or 0.5 V. (This is $4.0 / (2 \cdot 2^n)$, as we said previously in our definition of resolution.) So an input which is in the range 2.5 – 3.5 V will give a HIGH output only to comparator output C_2 , and our digital estimation will correspond to 3.0 V. Hence, the threshold for the upper comparator (its “–” input) should be set at 2.5 V. Similarly for the remaining comparators we work out the values which are given in the table below, where V_{est} is the digital estimate which corresponds to each state.

V_{in} range	Comparator	Threshold	V_{est}	$C_2 C_1 C_0$	$Q_1 Q_0$
2.5 – 3.5 V	C_2	2.5 V	3.0 V	111	11
1.5 – 2.5 V	C_1	1.5 V	2.0 V	110	10
0.5 – 1.5 V	C_0	0.5 V	1.0 V	100	01
-0.5 – 0.5 V	–	–	0.0 V	000	00

Using Ohm's and Kirchhoff's Laws, we arrive at the resistance ratios shown in Fig. 29 in order to achieve the desired comparator thresholds. All that remains is to determine the gate logic to convert the pattern of comparator output to a 2-bit digital output. Generalizing from the above, we see that we have agreement with our previous statements: For an n -bit ADC, we require $2^n - 1$ comparators, and the resolution is $\Delta V / 2^{n+1}$, where ΔV is the full range of analog input.

Successive Approximation ADCs

This technique is illustrated by Fig.30, which is also the one given for Lab 5. It uses a digital feedback loop which iterates once on successive clock cycles. The function of the *successive approximation register*, or SAR, is to make a digital estimate of the analog input based on the 1-bit output of the comparator. The current SAR estimate is then converted back to analog by the DAC and compared with the input. The cycle repeats until the “best” estimate is achieved. When that occurs, this present best estimate is latched into the output register (written into memory). By far the most common algorithm employed by SARs is the *binary search algorithm*. This is the one used by the SAR in Lab 5, and is illustrated in the example in the next section.

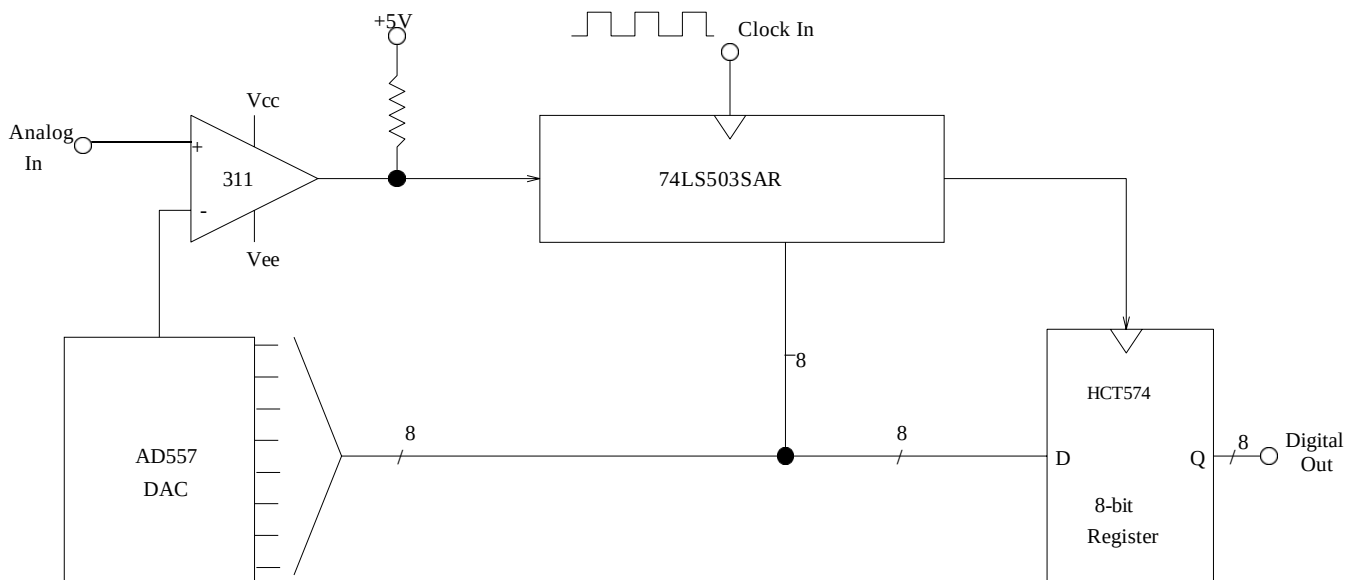


Figure 30: Scheme for 8-bit successive approximation, or feedback, ADC.

Binary Search Example

In this example we will see the binary search algorithm in action. The binary search algorithm can be summarized with the following words: *Go to the midpoint of the remaining non-excluded range*. In our example, we assume an 8-bit ADC with an expected input voltage range of 0 to 10V. So, naturally we choose the digital output to be $00000000_2 = 0$ when the input is 0V, and $11111111_2 = 255$ when the input is 10V. Hence, the LSB represents a voltage step $\Delta V = 10/255 = 39.22 \text{ mV}$.

Let the input voltage be some arbitrary value, 7.09 V. Now let's see how the algorithm works. Translating the words for the algorithm, written above, to what the SAR actually does is straightforward. The SAR always outputs one of two results, depending upon whether the output from the comparator was TRUE or FALSE. More precisely, the comparator will issue a HIGH if the current estimate is too small compared to the actual input, or a LOW if it is too big. The SAR then does the following:

1. If estimate too small, add 1 to MSB – $(n+1)$; or

2. If estimate too big, subtract 1 from MSB $-(n+1)$.

where n is the current clock cycle (see table).

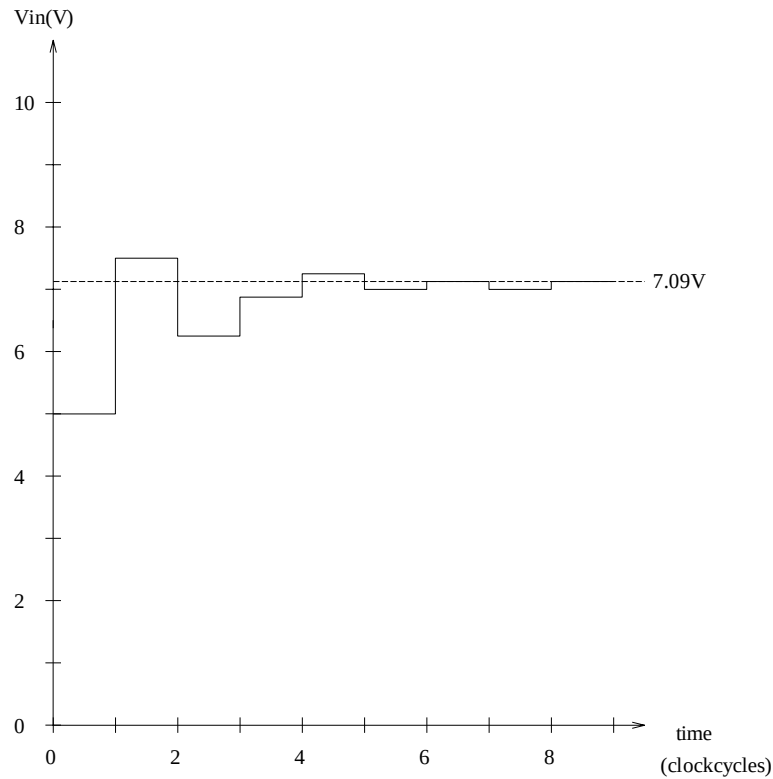


Figure 31: Binary search algorithm in action. The analog input is 7.09 V. The digital estimate for each clock cycle is represented by the solid line, and corresponds to the value of V_{est} in the table below.

Clock Cycle, n	SAR Bits	SAR Bit Sum	$V_{est}(V)$	comp. decision
0	01111111	127	4.98	toosmall
1	10111111	$127+64=191$	7.49	toobig
2	10011111	$191-32=159$	6.24	toosmall
3	10101111	$159+16=175$	6.86	toosmall
4	10110111	$175+8=183$	7.18	toobig
5	10110011	$183-4=179$	7.02	toosmall
6	10110101	$179+2=181$	7.10	toobig
7	10110100	$181-1=180$	7.06	toosmall

The binary search algorithm is guaranteed to find the best possible estimate in a number of clock cycles equal to the number of bits. In the example above, the best estimate was actually determined on the seventh clock cycle ($n=6$). But since the input value was between the digital estimates 180 and 181, there was no way for the ADC to determine which estimate was closer to the actual input value (without adding one more bit). Since the input can fall anywhere within 180 and 181 with the equal likelihood, there should be no

bias introduced with this method due to systematically choosing a digital estimate which is too small or too big. This is the desired outcome.

The binary-search algorithm is fast and efficient, and also has the advantage that it completes its estimation in a well determined number of clock cycles. Hence, the final digital result can always be latched after n clock cycles, where n is the number of bits. (Many ADCs actually wait one additional clock cycle in order to guarantee that bits have settled, are latched properly, and are reset for the next input.)

Single/Dual Slope ADCs

These techniques are slower than flash or successive approximation, but in principle can be quite accurate. The improved accuracy is for two reasons, because time, which is robustly measured using digital techniques, is used as the measured quantity, and because there is some immunity to noise pickup, especially for the dual slope case.

The single slope technique is illustrated in Fig. 32, which is taken from Figure 9.54 of the text. The device near the input and the capacitor is an FET transistor which is used as a switch. When the input to the FET gate, which comes from the Q output of the D-type flip-flop, is LOW, then the FET is switched off, and it draws no current. However, when Q goes HIGH, the FET pulls the +input of the comparator to ground, and holds it there. The box marked "osc" represents a typical digital clock. The arrow within the circle connected to $+V_{cc}$ is the symbol for a "current source", which means that its output is a constant current, regardless of the impedance at its output (within reasonable bounds).

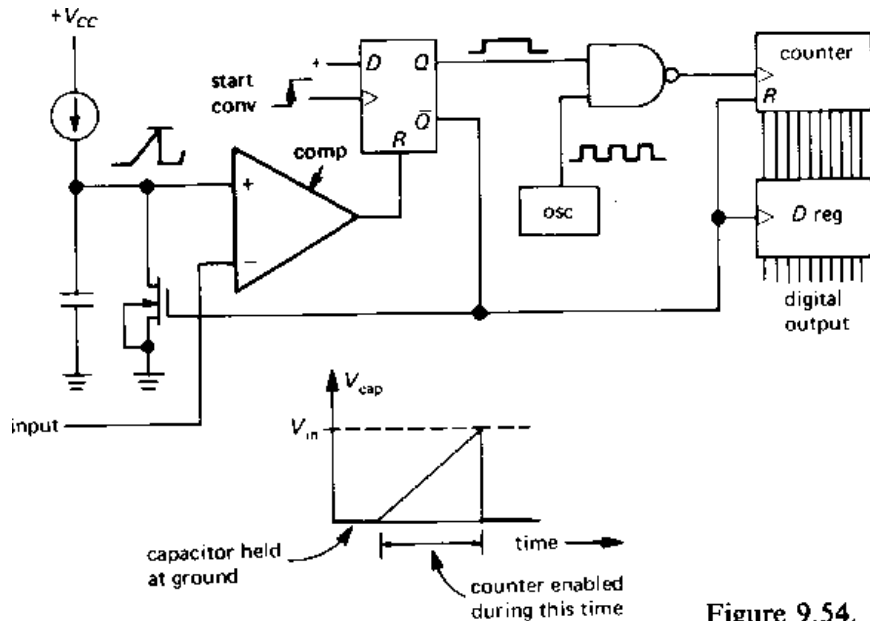


Figure 9.54. Si

Figure 32: Scheme for single-slope ADC, from text.

The process begins when a rising-edge signal is sent to the flip-flop, for example from a debounced switch. Since the D input is HIGH, then Q goes HIGH. Hence the counter, no longer being held at reset by the flip-flop, begins counting. At the same time the FET is

switched off and a signal is sent to the – input of the comparator. Now we must analyze the nature of this signal.

The voltage across a capacitor V_{cap} is related to its stored charge by $V_{cap} = Q/C$, where C is the capacitance. Differentiating gives $dV_c/dt = I/C$. Now, because of the current source, the right-hand side of this equation is a constant. Finally, since one side of the capacitor is at ground, then the comparator + input is just V_{cap} . Hence, we can integrate our expression over a time interval Δt to give:

$$V_+ = V_{cap} = (I/C)\Delta t$$

Since I/C is a known constant, this equation allows us to convert the V_+ input to a time Δt to be measured by the counter. This linear relation between $V_+ (=V_{cap})$ and Δt is illustrated in the figure. The counter stops (is reset) and its final count stored in the register when V_+ becomes equal to V_{in} , thus changing the state of the comparator. This also resets the flip-flop, thus returning the circuit to its initial state.

The dual-slope ADCs work similarly, but with a two-step process. First, a capacitor is charged for a fixed time τ with a current source whose current is proportional to V_{in} , $I = \alpha V_{in}$, where α is the constant of proportionality. Hence, V_{cap} is proportional to τ : $V_{cap} = \alpha V_{in} \tau / C$. The capacitor is then discharged at constant current I' and the time Δt to discharge is measured. Therefore,

$$\Delta t = [C/I'] [\alpha \tau / C] V_{in} = \beta V_{in}$$

where $\beta = \alpha \tau / I'$ is a known constant.

This technique has two advantages compared with single-slope. First, we see from the equation above that the result is independent of C . This is good, as precise capacitance values are difficult to fabricate. Second, the integration of the input voltage in the charge-up step allows 60 Hz pickup noise (or other periodic noise) to be averaged to zero.

6 Counters, Registers, and State Machines II

The general scheme for a *state machine* is given in Fig.33. It has n bits of memory, k inputs, and m outputs. It consists of a synchronous data register (lower box) which stores the machine's *present state*. A set of separate flip-flops can be used for this, as long as they are clocked synchronously. The logic in the upper box acts upon the current state, plus any inputs, to produce the machine's next state, as well as any outputs. Upon each pulse of the clock input CLK, the machine is moved from the present state to the next state. We will introduce this topic using counters as examples, then moving to more general applications. We will see, in fact, that the state machine represents a simple *processor*: The inputs can be generalized to be the processor program and the logic might be replaced by a random-access memory (RAM).

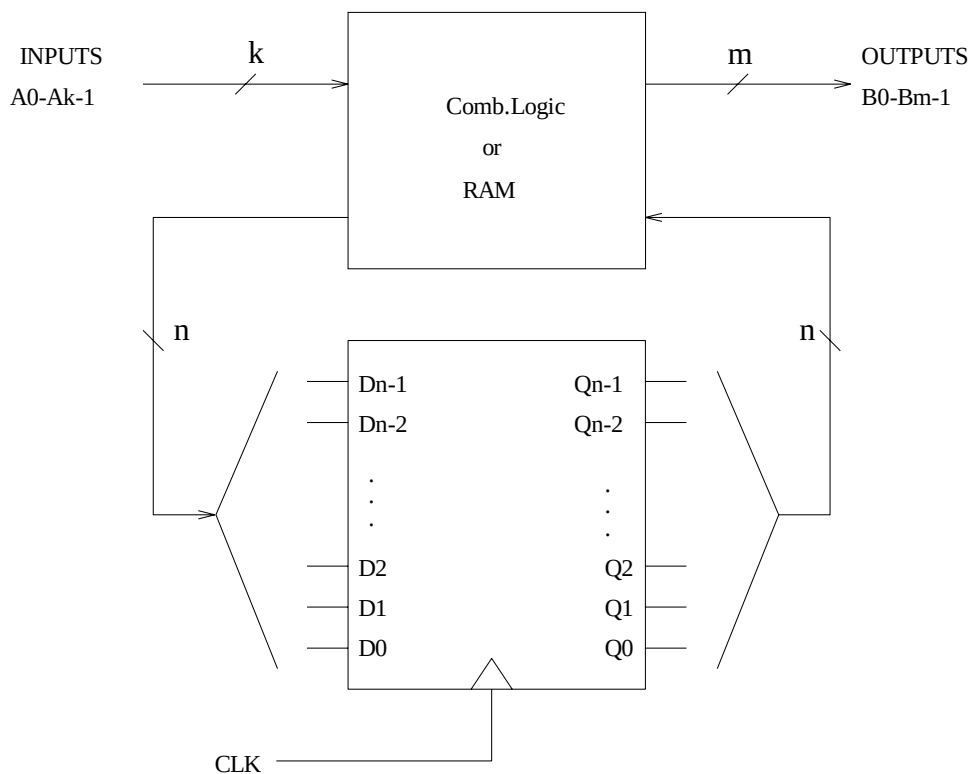


Figure 33: General scheme for state machine.

The strategy for applying this scheme to a given problem consists of the following:

1. Identify the number of required states, I . The number of bits of memory (e.g. number of flip-flops) required to specify the m states is at minimum $n = \log_2(m)$.
2. Make a *state diagram* which shows all states, inputs, and outputs.
3. Make a truth table for the logic section. The table will have $n+k$ inputs and $n+m$ outputs.
4. Implement the truth table using our combinational logic techniques.

State Machine Introduction: Synchronous Counters

Counters implemented as state machines are always *synchronous*, that is the entire circuit is in phase with the clock. Recall that our previous “ripple” counters were asynchronous — logic was initiated at different times throughout the circuit. Synchronous systems are essential whenever a sequential system requires more than a very modest speed or complexity.

Example: Up/down 2-bit Synchronous Counter

A 2-bit counter requires 4 states, with each state corresponding to one of the 4 possible 2-bit numbers. Hence, 2 bits of memory are required. We will use 2 flip-flops (D-type) to implement this. The state diagram is given in Fig. 34. Each circle represents one of the states, and the arrows represent a clock pulse which offers a transition to another state (or possibly to remain at the present state). The 4 states are specified by the 2 bits of memory: $A = 00, B = 01, C = 10, D = 11$. Note that we are free to label the states as we choose, as long as they are uniquely specified. However, in this case it is easiest to choose labels which correspond to our desired outputs, that is the 2-bit binary sequence 00, 01, 10, and 11. Hence, these labels are equivalent to our desired outputs, call them Q_1Q_0 , which are available at each state. (Note that the lettered labels $A-D$ are superfluous; they could be omitted.)

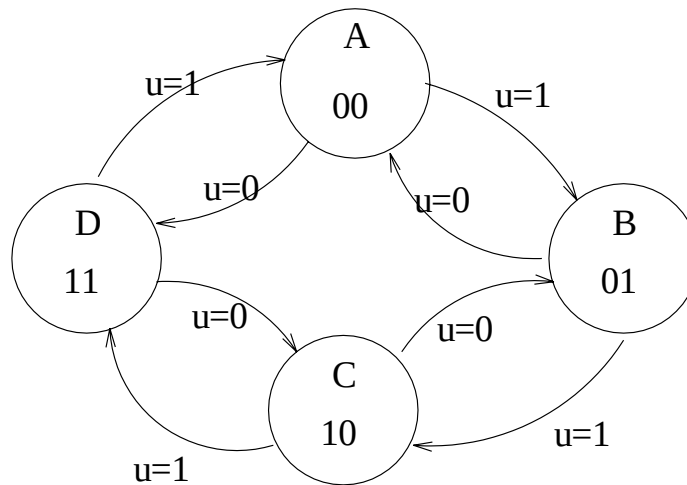


Figure 34: State diagram for 2-bit up/down synchronous counter.

Our processor has one input bit u , which programs the up-counting ($u=1$) or down-counting ($u=0$) functions. In this case, the state machine outputs are the two bits of the present state, Q_1Q_0 , so we do not reproduce them in our truth table. The truth table for the logic is below.

	PresentState		NextState	
u		Q_1Q_0		D_1D_0
1	A	00	B	01
1	B	01	C	10
1	C	10	D	11
1	D	11	A	00
0	A	00	D	11
0	D	11	C	10
0	C	10	B	01
0	B	01	A	00

We can now invoke the logic as usual. We have 2 “outputs”, D_0 and D_1 , which are to be evaluated separately. From the truth table, or using a K-map, we see that

$$D_1 = u \oplus (Q_0 \oplus Q_1); \quad D_0 = Q_0$$

Example: Divide-by-Three Synchronous Counter

Our state machine is supposed to count input pulses (input at the CLK) and set an output bit HIGH on every 3rd input pulse. Note that this could represent either a 2-bit (total) counter, or more generally the 2 least-significant bits of a many-bit counter.

We require 3 states, therefore we need 2 bits of memory (2 D-type flip-flops, for example). These 2 flip-flops can describe 4 states, so we will have one “unused” state. A state diagram is shown in Fig. 35, with one way of labelling the states and output bit (called p) given.

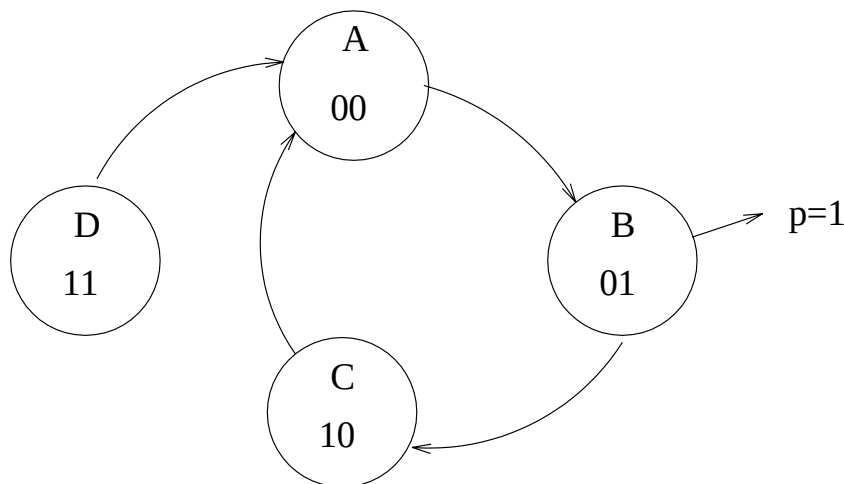


Figure 35: State diagram for a divide-by-3 synchronous counter.

The truth table for the combinational logic is below. It is important that the “extra state” $D=11$ be given an exit path, otherwise your processor may end up there upon power-up and remain stuck. (This effect has probably come to your attention with the “frozen” computer, which may require a reboot.) Also, note that we could have taken the output p from any of the states A, B , or C .

PresentState		NextState		Output
Q_1Q_0		D_1D_0		p
A	00	B	01	0
B	01	C	10	1
C	10	A	00	0
D	11	A	00	0

What are the logic expressions for our 3 “outputs” of this truth table (D_1 , D_2 , and p)? How would this be implemented with D-type flip-flops and logic gates? With JK flip-flops replacing the D-type?

7 Memories and Processors

Memory Terminology

We will not discuss the topic of data storage technologies *per se*. We are mostly interested here in the question of how data storage can be effectively *organized*. The important common element of the memories we will study is that they are *random access memories*, or RAM. This means that each bit of information can be *individually* stored or retrieved — with a valid input address. This is to be contrasted with *sequential* memories in which bits must be stored or retrieved in a particular sequence, for example with data storage on magnetic tape. Unfortunately the term RAM has come to have a more specific meaning: A memory for which bits can both be easily stored or retrieved (“written to” or “read from”). Here is a rundown on some terms:

- RAM. In general, refers to random access memory. All of the devices we are considering to be “memories” (RAM, ROM, etc.) are random access. The term RAM has also come to mean memory which can be both easily written to and read from. There are two main technologies used for RAM:

1.) Static RAM. These essentially are arrays of flip-flops. They can be fabricated in ICs as large arrays of tiny flip-flops.) “SRAM” is intrinsically somewhat faster than dynamic RAM.

2.) Dynamic RAM. Uses capacitor arrays. Charge put on a capacitor will produce a HIGH bit if its voltage $V = Q/C$ exceeds the threshold for the logic standard in use. Since the charge will “leak” off through the resistance of the connections in times of order ~ 1 msec, the stored information must be continuously refreshed (hence the term “dynamic”). Dynamic RAM can be fabricated with more bits per unit area in an IC than static RAM. Hence, it is usually the technology of choice for most large-scale IC memories.

- ROM. *Read-only memory*. Information cannot be easily stored. The idea is that bits are initially defined and are never changed thereafter. As an example, it is generally prudent for the instructions used to initialize a computer upon initial power-up to be stored in ROM. The following terms refer to versions of ROM for which the stored bits *can* be over-written, but not easily.
- PROM. *Programmable ROM*. Bits can be set on a programming bench by burning “fusible links,” or equivalent. This technology is also used for programmable array logic (PALs), which we will briefly discuss in class.
- EPROM. ROM which can be erased using ultraviolet light.
- EEPROM. ROM which can be erased electronically. A

few other points of terminology:

- As you know, a bit is a binary digit. It represents the smallest element of information.
- A byte is 8 bits.
- A “K” of memory is $2^{10} = 1024$ bits (sometimes written KB). And a megabit (MB) is $1K \times 1K$ bits.

- RAM is organized into many data “words” of some prescribed length. For example, a RAM which has $8K=8192$ memory locations, with each location storing a data word of “width” 16 bits, would be referred to as a RAM of size $8K \times 16$. The total storage capacity of this memory would therefore be 128KB, or simply a “128K” memory. (With modern very large scale integration (VLSI) technology, a typical RAM IC might be ~ 16 MB.)
- Besides the memory “size,” the other important specification for memory is the *access time*. This is the time delay between when a valid request for stored data is sent to a memory and when the corresponding bit of data appears at the output. A typical access time, depending upon the technology of the memory, might be ~ 10 ns.

Memory Configuration

As stated above, the term “memory” refers to a particular way of organizing information — by random access — which is distinct from the less specific term “data storage.” Figure 36 shows how an 8-bit RAM (8×1) is organized. (This is a very small memory, but illustrates the concepts.) Our RAM consists of three main components: an 8-bit multiplexer, an 8-bit demultiplexer, and 8 bits of storage. The storage shown consists of edge-triggered D-type flip-flops. Hence, this is evidently a “static RAM.” (There is no fundamental reason for using edge-triggered flip-flops. They could just as easily be level-triggered, like the simple “clocked” S-R flip-flop of Fig. 14.)

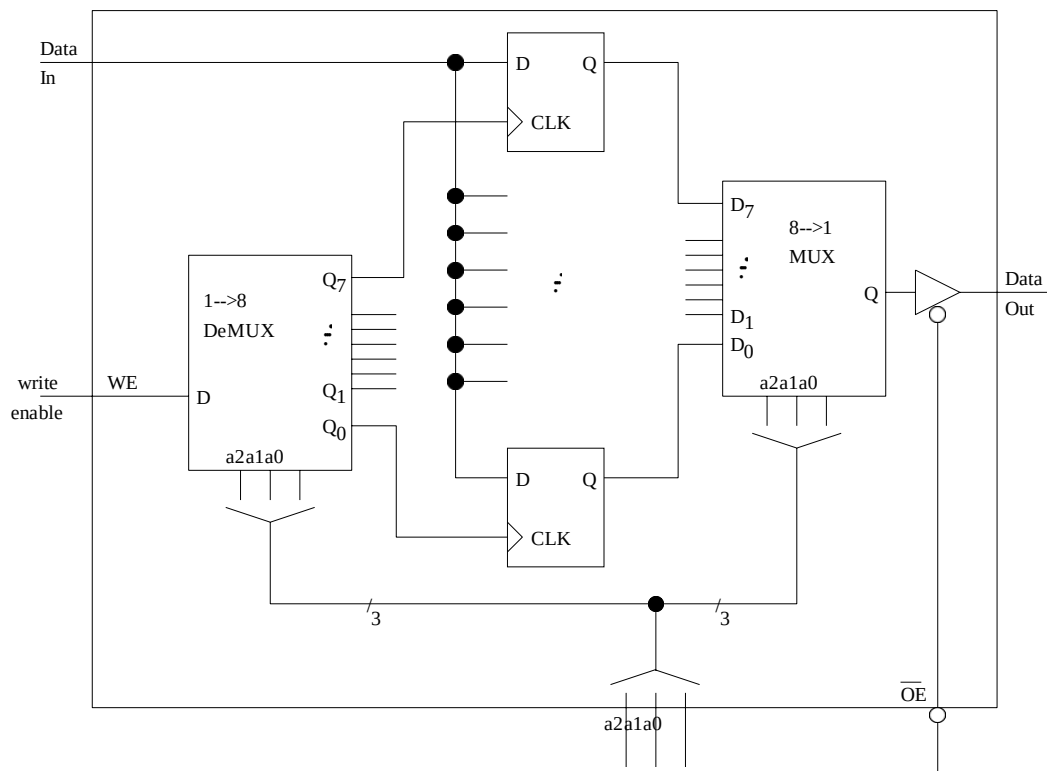


Figure 36: An 8×1 bit RAM.

Our example RAM has 6 external connections which are inputs (data in, write enable (WE), 3-state enable ($\overline{OE}$), and 3 address bits ($A = a_2a_1a_0$), and has one output connection (data out), giving 7 external connections total, plus 2 for power/ground. To write information to the RAM, one would supply a valid address, for example $A=101$. The data bit to be written to location 101 is to appear at the data input as either a logic HIGH or LOW signal. And to enable the writing into this bit, the WE signal must be asserted. This then appears at the Q_5 output of the demultiplexer, and is passed on to the appropriate flip-flop, which stores the input data bit and passes it on to the Q_5 multiplexer input.

To read data from our RAM, one asserts an address, so that the selected bit is sent to the MUX output and then the 3-state buffer. The purpose of the 3-state buffer is to ensure that no digital outputs are directly connected together, for example if our RAM output were connected to a data "bus," which in turn was connected to several other devices. Recall that the 3-state devices have outputs which are effectively disconnected if there is no enable signal. So if the output data connection of our RAM is connected to a data bus, then the OE signal must be coordinated with any other outputs also connected to the data bus. When it is OK to read data from the RAM (all other output devices are disconnected from the bus), the $\overline{OE}$ signal is asserted and the MUX output will appear at the RAM output.

One could of course also store the 8 bits of data directly to an 8-bit data register, rather than using the RAM configuration outlined above. In this case, the number of external connections is 17 (8 data in, 8 data out, and 1 clock), compared with the 7 of our RAM. For a more realistic case where the number of bits of memory n is much larger than our example, we generalize the above to arrive at $4 + \log_2(n)$ external connections for the RAM, compared with $1 + 2n$ for the standalone register. Obviously for large n , the register is impractical, whereas the RAM remains reasonable. Actually, it is even somewhat better than this for the RAM case, since the number of external connections does not grow with the width of the stored data words. Hence, a RAM of size $1K \times 16 = 16KB$ requires only 14 connections. This is to be compared with 32,001 connections for the register. Note that the RAM can only supply one bit at a time to the output. This may seem like a handicap, but is actually well matched to standard microprocessors.

A State Machine with Memory

For reference, our usual state machine configuration is shown again in Fig. 37. Now we consider the use of a memory with a state machine, as depicted in Fig. 38. A random access memory is used in place of the usual combinational logic. (A ROM has been specified, to emphasize that we are not changing the memory—once it is defined initially, it is only read from. The memory is used to conveniently encode the connection between present and next states.

To start with, let's assume a state machine with no external inputs or outputs. Then the state machine's present state (PS) becomes an *address* which is input to the ROM. The *data word* stored in the ROM at that address then corresponds to the next state (NS). This correspondence had been initially programmed into the ROM, just as the specific combinational logic in our old state machine had to be pre-determined. So if the PS as defined by the Q bits at the data register are, for example, 1001, then the ROM data word at address 1001 will be the NS which is then passed back to the register. When there are also external inputs, as there will be for most anything of interest, these are combined with the PS bits to form a longer address for the ROM. Similarly, any external outputs are recombined with

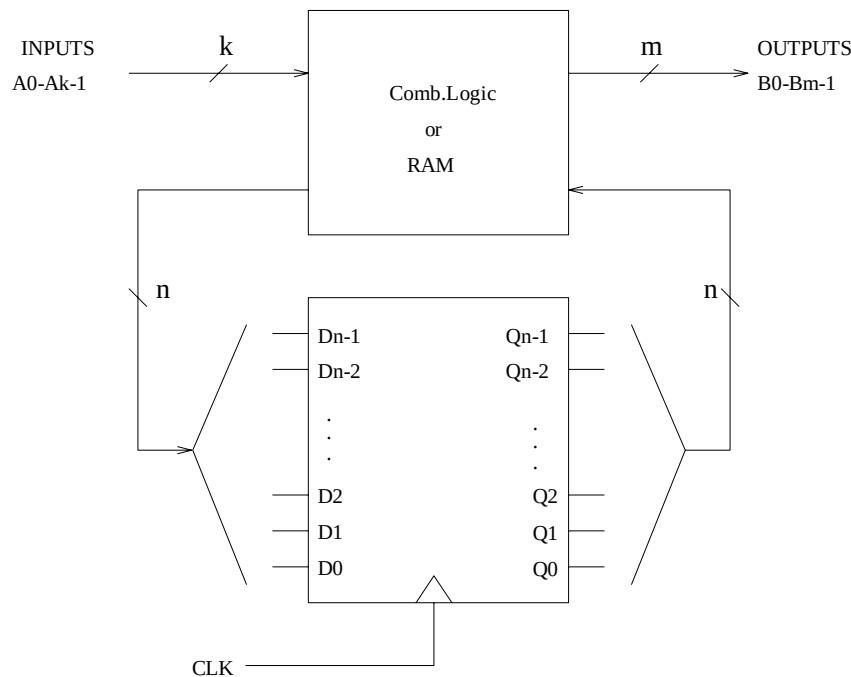


Figure37:Thestandardstatemachineconfiguration.

theNSbitsinthedataword.

This should become clear with an example.

Example: Divide by 2 or 3 Counter

We will use a state machine with ROM, as in Fig. 38, to design a counter which either divides by 2 or by 3, depending upon the value of an external input bit p . This state machine will require 3 states, therefore we will need to describe 4 states, using 2 bits. We can label the states $A = 00$, $B = 01$, $C = 10$, and $D = 11$. Let $p = 0$ be the divide by 2 case, and $p = 1$ the divide by 3. The output bit $r = 1$ when the present state is B . Otherwise $r = 0$. State D is normally unused. The truth table is below. The student should draw the corresponding state diagram.

p	Present State		Next State		r
	$Q_1 Q_0$		$D_1 D_0$		
0	00	A	B	01	0
0	01	B	A	00	1
0	10	C	A	00	0
0	11	D	A	00	0
1	00	A	B	01	0
1	01	B	C	10	1
1	10	C	D	11	0
1	11	D	A	00	0

This ROM requires 3 address bits (2 for PS and 1 for input bit p), which corresponds to 8 locations in memory. Each location has a data word which has length 3 bits (2 for NS and

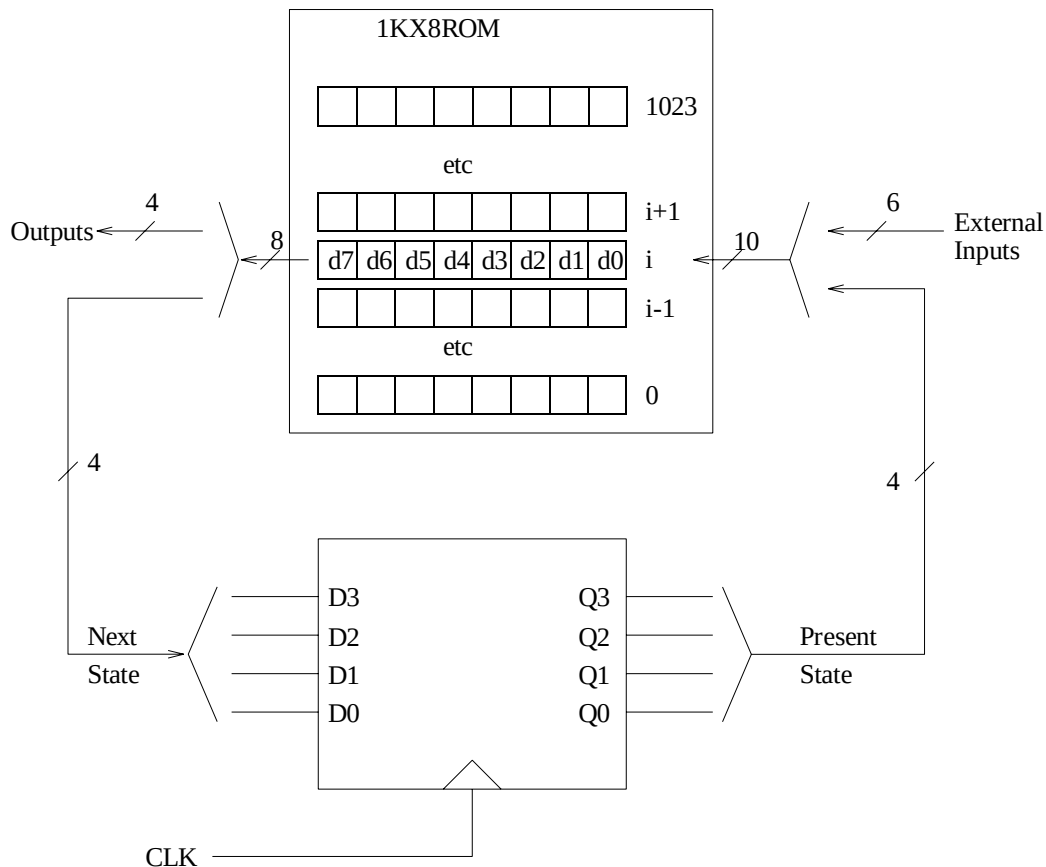


Figure38:Toward microprocessor:Replacing the combinational logic with a memory.

1 for the output bit r). Therefore, the size of this memory is 8×3 , or 24 total bits. A very small ROM indeed. The programming of the ROM is very straightforward and can be read directly from the truth table above. We just need to set an encoding convention. Let the addresses be encoded as pQ_1Q_0 and the data words as D_1D_0r . For example, let's look at the 5th row of the truth table. The address would be 100 and the data word at this address would be 010. The remaining bits of the ROM would be programmed in the same way. So one would initially "burn in" these bit patterns into the ROM and put it into the circuit. That's all there is to it. Of course if one were careful not to overwrite the memory, or if an evolving logical pattern were required, then a RAM could be used instead of the ROM.

Generalization to Microprocessors

A state machine with zero input bits can perform a counter-like function, but not more: its next state is limited to be a function only of the present state. A single input bit can be used to "program" the state machine to behave in one of two possible ways for each present state, as we discussed, for example, with the up/down counter of Section 4.4.1, or the example in the preceding section. On the other hand, with n inputs, the machine can perform 2^n different operations. So, for example, with $n = 8$ the machine can perform one of 256 different operations on each clock cycle. This tremendous potential and flexibility. The input bits can themselves be sequenced — stored externally in a specific sequence which

is then applied step by step to the state machine inputs on successive clock cycles. Such a stored sequence of operations is a *program* and the 256 operations represent the programming operations. In Fig. 38 we have essentially configured a simple *microprocessor*. The inputs and outputs would need to be connected to buses (via 3-state buffers where appropriate), which in turn are also connected to memories which store the program and any output or input data. The buses would also be connected to various input/output devices, mass storage devices, etc.