

Discrete - Time Signals and Systems

FIR Filters-I

Yogananda Isukapalli

Discrete-Time System

Something that can transform an input sequence into an output sequence of numbers

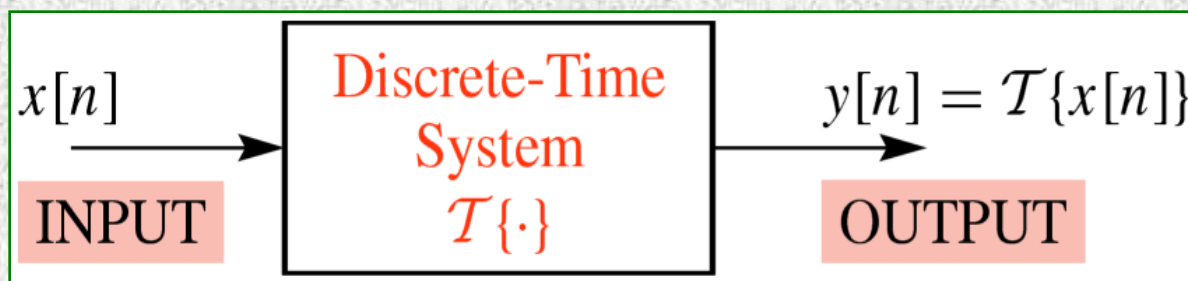
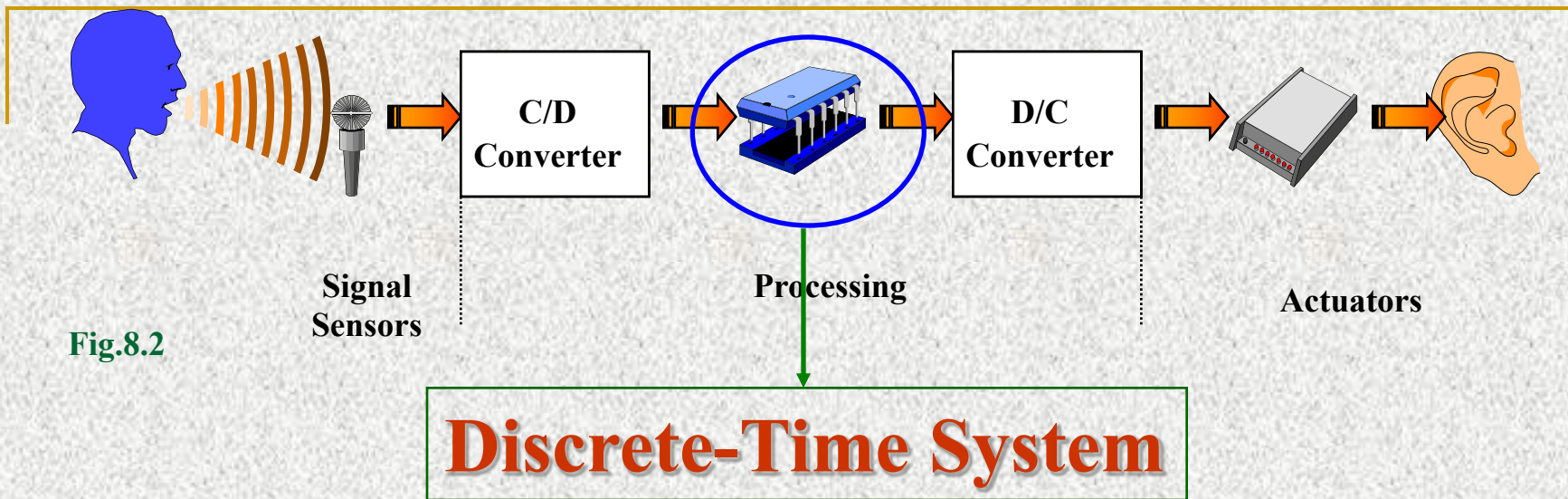


Fig.8.1

Since a discrete-time signal is a sequence of numbers, the operator \mathcal{T} can be described by a mathematical formula. It is just a computational process



An Evolution of
computing
machines

For the price of a
small house, you
could have one
of these

**Model 4136
PROGRAMMABLE
DIGITAL
FILTER**

Variable-Order Digital Filter for Realizing All Classical Designs

The Rockland Model 4136 Programmable Digital Filter consists of a second-order digital filter section which is multiplexed four ways to achieve eighth-order filtering. Each of the four sections has fully-programmable coefficients which are stored internally in a read-write memory.

Filter input and output words are in 16-bit parallel form at a maximum sampling rate of 80 KHz while internal computations are made with 24-bit accuracy.

TRANSFER FUNCTION

The transfer function from filter input to filter output in z-transform notation is given by

$$H_N(z) = \prod_{n=1}^N \frac{K_n(1+z^{-1}A1+z^{-2}A2)}{1-z^{-1}B1-z^{-2}B2} \quad (1)$$

where N=0,1,2,3,4 is one-half the filter order se-

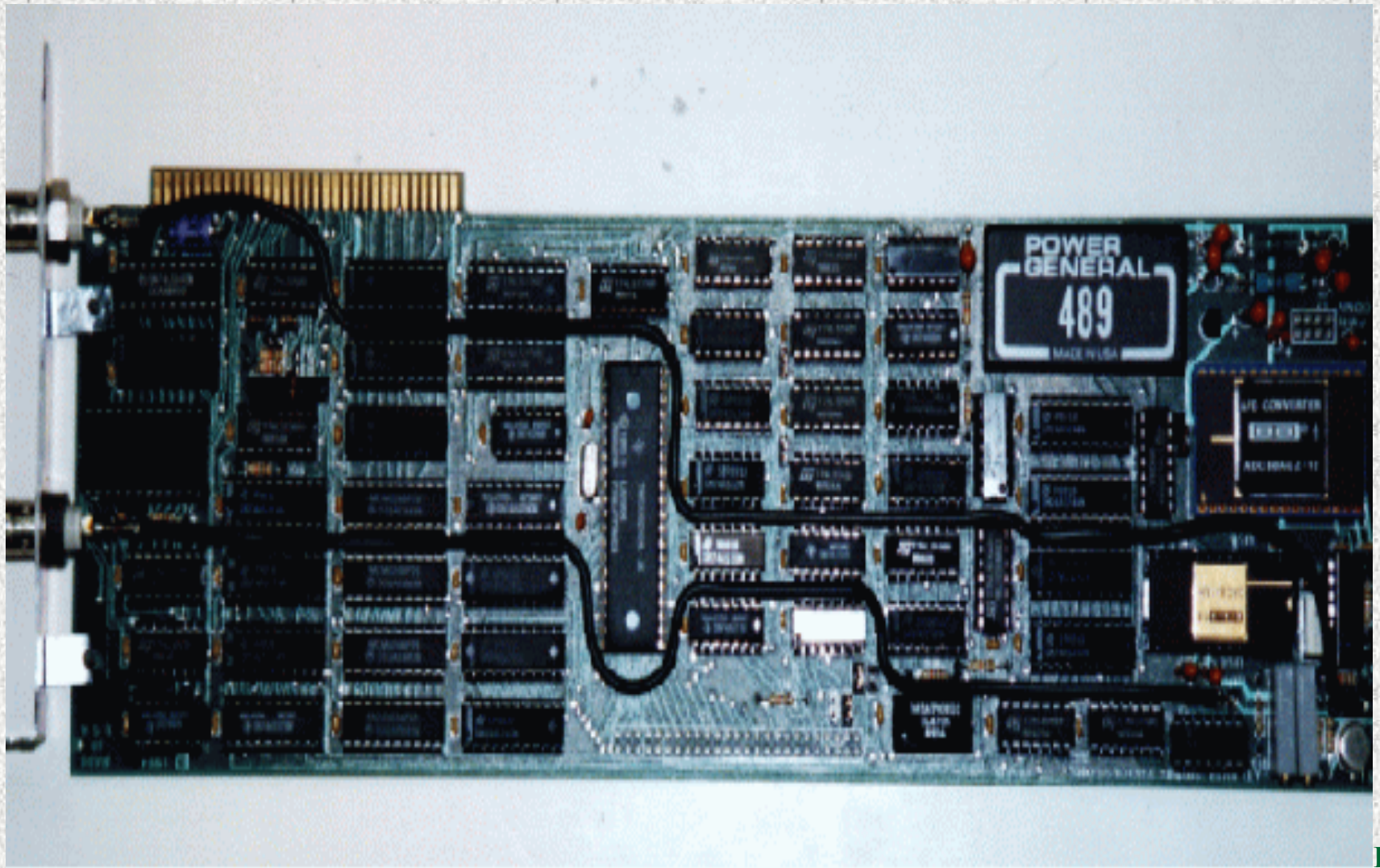
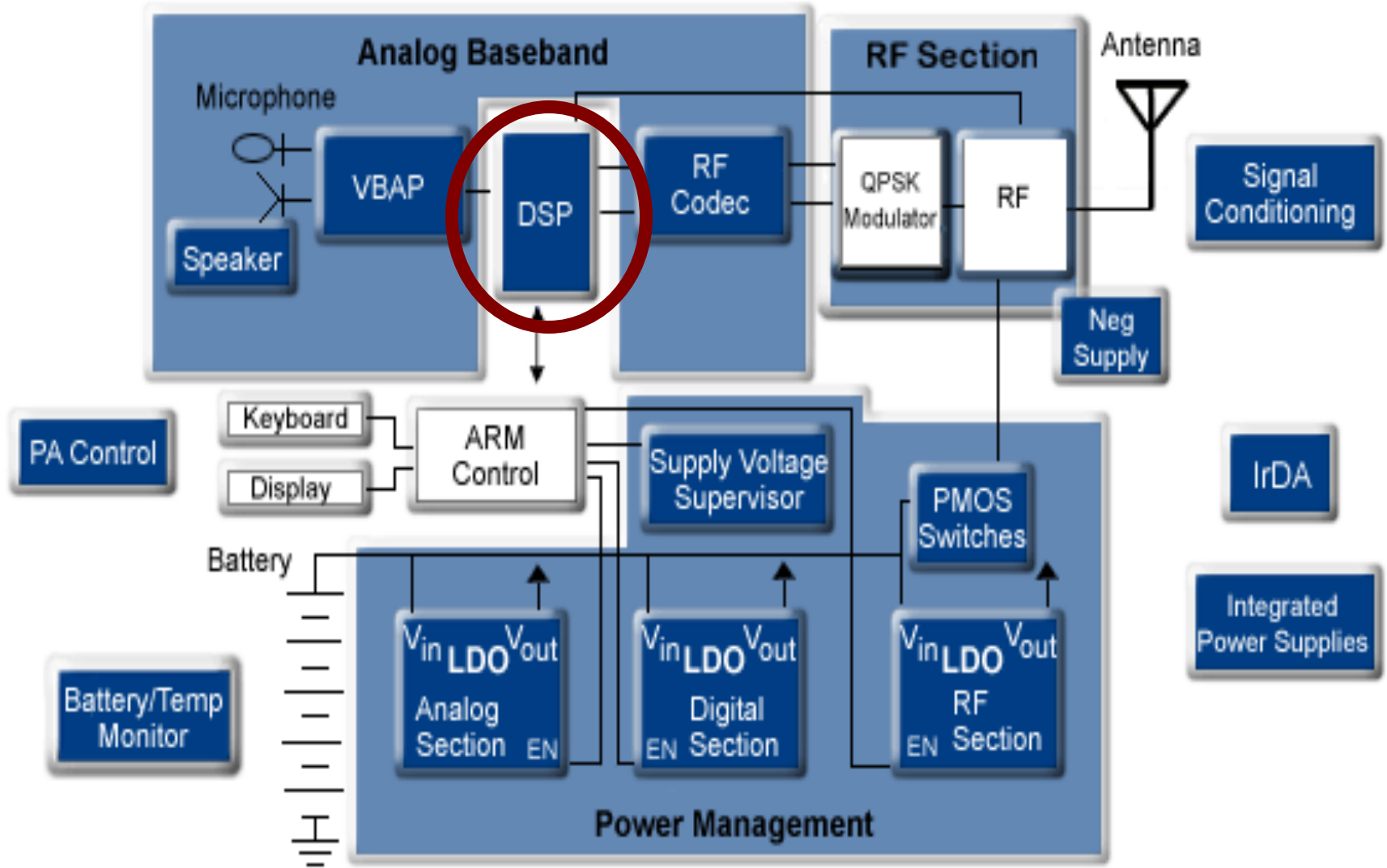


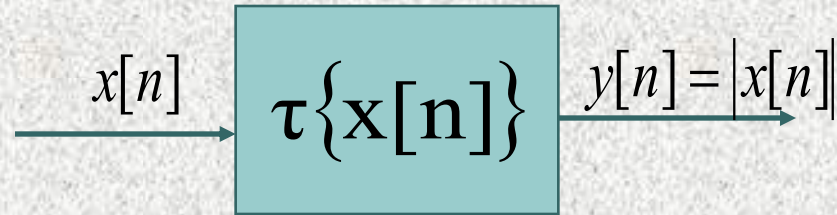
Fig.8.4

TMS32010, 1983: First PC plug-in board from Atlanta Signal Proc

Digital Cell Phone (ca. 2000)



Example 1: A Simple Discrete System



Output is the absolute value of input

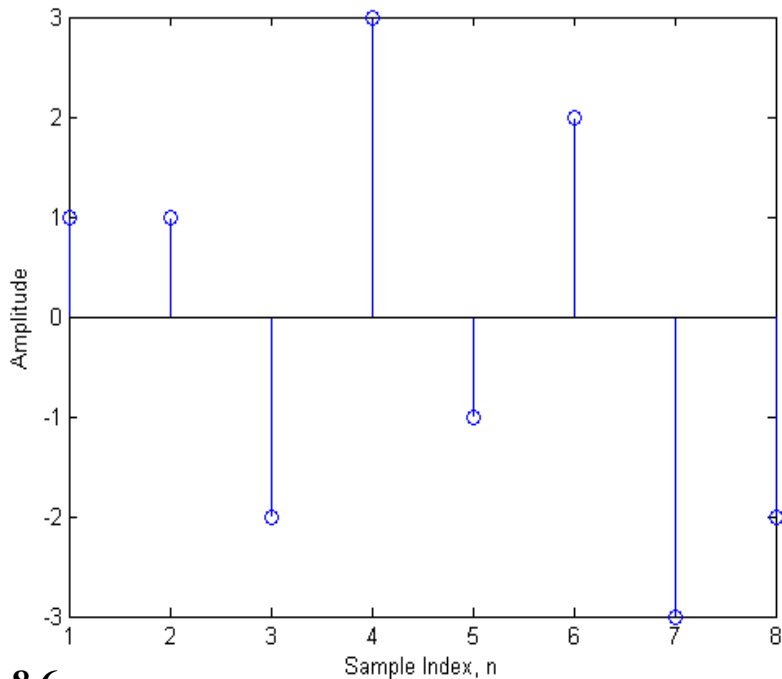


Fig.8.6

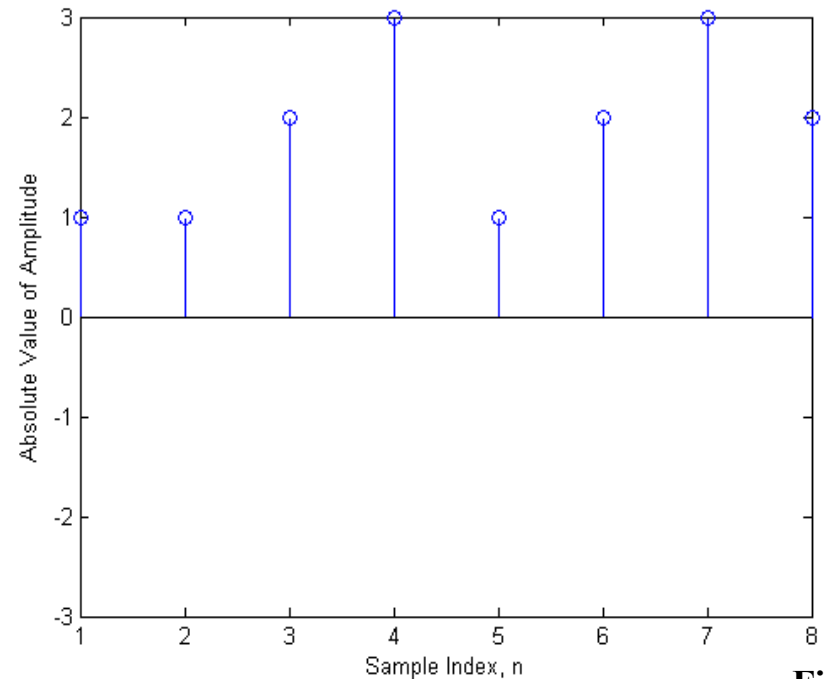
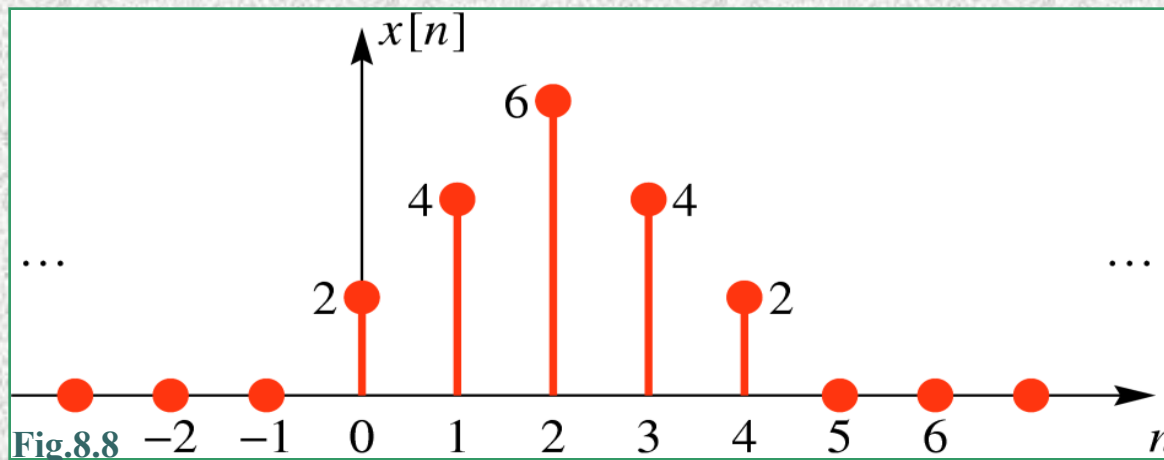


Fig.8.7

Example2: 3-point averaging method

Consider an input sequence $x[n]$



$$x[0] = 2 \quad x[3] = 4$$

$$x[1] = 4 \quad x[4] = 2$$

$$x[2] = 6 \quad x[5] = 0$$

Take the average of
any three points in a
sequence

$$\frac{\{2 + 4 + 6\}}{3} = 4, \quad \frac{\{4 + 6 + 4\}}{3} = 14/3$$

$$\frac{\{6 + 4 + 2\}}{3} = 4, \quad \frac{\{4 + 2 + 0\}}{3} = 2$$

Notation for the output is arbitrary, following the notation shown below leads to the output;

$$y[0] = \frac{\{x[0] + x[1] + x[2]\}}{3} \quad y[1] = \frac{\{x[1] + x[2] + x[3]\}}{3}$$

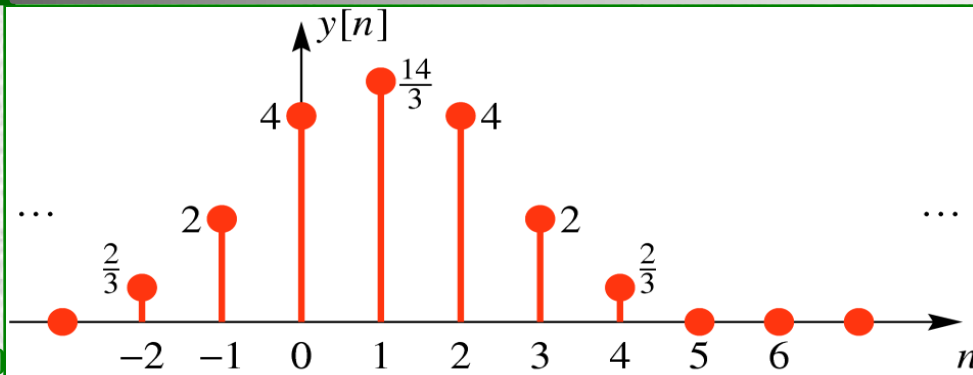
n	$n < -2$	-2	-1	0	1	2	3	4	5	$n > 5$
$x[n]$	0	0	0	2	4	6	4	2	0	0
$y[n]$	0	$\frac{2}{3}$	2	4	$\frac{14}{3}$	4	2	$\frac{2}{3}$	0	0

$n=0$ $y[0] = \frac{1}{3}(x[0] + x[1] + x[2])$

$n=1$ $y[1] = \frac{1}{3}(x[1] + x[2] + x[3])$

Fig.8.9

Output



The above process (3-point average) generalizes to an important input-output equation known as *difference equation*

$$y[n] = \frac{1}{3}(x[n] + x[n+1] + x[n+2])$$

The above equation describes a very important class of discrete-time systems called '*FIR filters*'

However, the equation $y[n] = \frac{1}{3}(x[n] + x[n+1] + x[n+2])$
doesn't seem practical as we need two future
samples to calculate present output

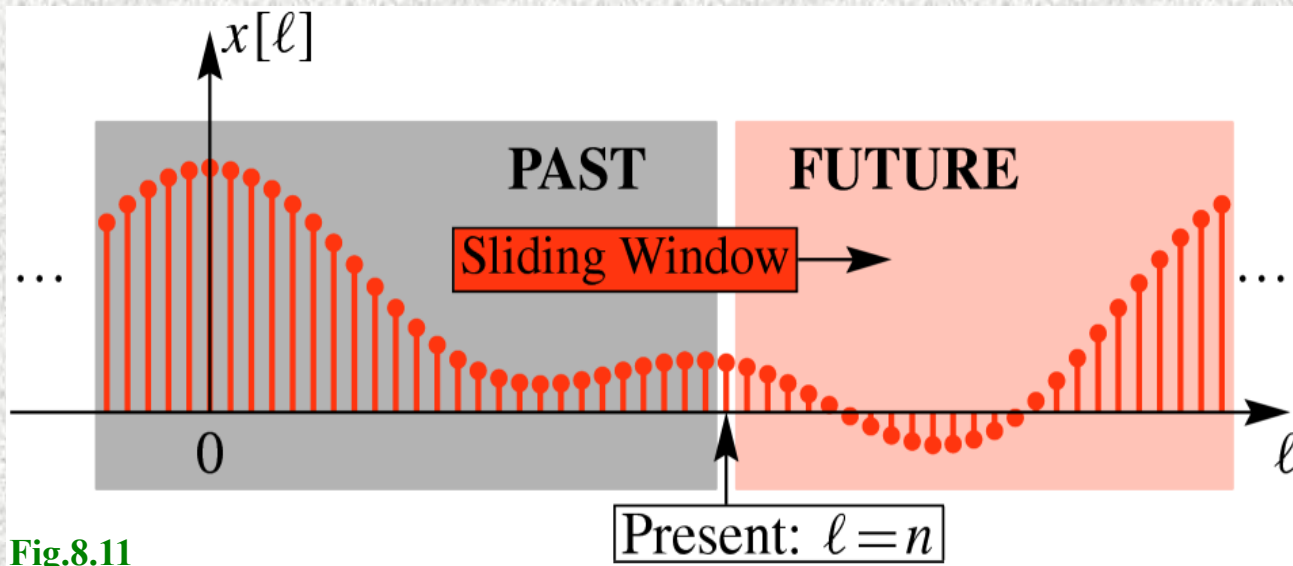


Fig.8.11

$$y[n] = \frac{1}{3}(x[n] + x[n-1] + x[n-2])$$

Depends only on past
values, a causal system

Difference equations

$$y[n] = \sum_{l=1}^N a_l y[n-l] + \sum_{k=0}^M b_k x[n-k], \quad \text{Recursive equation}$$

Represents a class of filters known as IIR filters

$$y[n] = \sum_{k=0}^M b_k x[n-k], \quad \text{Non-Recursive part}$$

FIR filters

First order discrete – time system

$$y[n] = a_1 y[n-1] + b_0 x[n] \quad \text{a Recursive equation}$$

Solution :

$$n = 1 \quad y[1] = a_1 y[0] + b_0 x[1]$$

$$n = 2 \quad y[2] = a_1 y[1] + b_0 x[2]$$

$$= a_1 (a_1 y[0] + b_0 x[1]) + b_0 x[2]$$

$$= a_1^2 y[0] + a_1 b_0 x[1] + b_0 x[2]$$

$$n = 3 \quad y[3] = a_1 y[2] + b_0 x[3]$$

$$= a_1 (a_1^2 y[0] + a_1 b_0 x[1] + b_0 x[2]) + b_0 x[3]$$

$$y[3] = a_1^3 y[0] + a_1^2 b_0 x[1] + a_1 b_0 x[2] + b_0 x[3]$$

Generalizing the 1st order discrete-time system,

$$y[r] = a_1^r y[0] + b_0 [a_1^{r-1} x[1] + a_1^{r-2} x[2] + \dots + a_1^0 x[r]]$$

$$y[r] = a_1^r y[0] + b_0 \sum_{m=1}^r a_1^{(r-m)} x[m]$$

For a causal system, $y[0] = a_1 y[\cancel{-1}] + b_0 x[0] = b_0 x[0]$

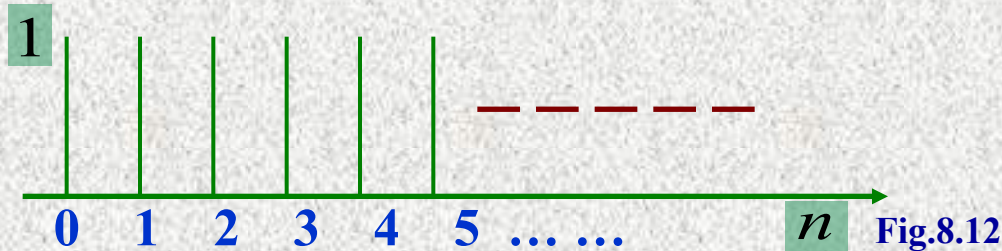
$$\therefore y_c[r] = a_1^r b_0 x[0] + b_0 \sum_{m=1}^r a_1^{(r-m)} x[m]$$

$$y_c[r] = b_0 \sum_{m=0}^r a_1^{(r-m)} x[m]$$



Example

Let $x[n] = 1$ for all n , $n \geq 0$



$$y[r] = a_1^r y[0] + b_0 \left[a_1^{r-1} x[1] + a_1^{r-2} x[2] + \dots + a_1^0 x[r] \right]$$

$$= a_1^r y[0] + b_0 \left[a_1^{r-1} + a_1^{r-2} + \dots + 1 \right]$$

$$y[r] = a_1^r y[0] + b_0 \frac{a_1^r - 1}{a_1 - 1} \quad a_1 \neq 1$$

$$y[r] = y[0] + b_0 r \quad a_1 = 1$$

$$\begin{aligned} \therefore \lim_{a_1 \rightarrow 1} \frac{a_1^r - 1}{a_1 - 1} &= \lim_{a_1 \rightarrow 1} \frac{\frac{d}{da_1} (a_1^r - 1)}{\frac{d}{da_1} (a_1 - 1)} \\ &= \lim_{a_1 \rightarrow 1} \frac{r a_1^{r-1}}{1} = r 1^{r-1} = r \end{aligned}$$

Let $a_1 = 1/2$, $b_0 = 2$, $y[0] = 3$

$$\begin{aligned} y[n] &= (1/2)^n 3 + 2 \frac{(1/2)^n - 1}{(1/2) - 1} \\ &= (1/2)^n 3 - 4 \left[(1/2)^n - 1 \right] = 4 - (1/2)^n \quad n \geq 0 \end{aligned}$$



The General FIR Filter

Non - Recursive part of difference equation represents a general FIR filter

$$y[n] = \sum_{k=0}^M b_k x[n - k],$$

The above equation doesn't involve any past samples, so the system is a *causal* one. The moving average problem discussed earlier is an FIR filter



Filter Order = M: *No. of memory blocks required in the filter implementation*

Filter Length, $L = M+1$: *Total No. of samples required in calculating the output, M from memory (past) and one present sample*

Filter coefficients $\{b_k\}$: *Completely define an FIR filter. All the properties of the filter can be understood through the coefficients*

Graphical view of a general FIR filter

$$y[n] = \sum_{k=0}^M b_k x[n - k]$$

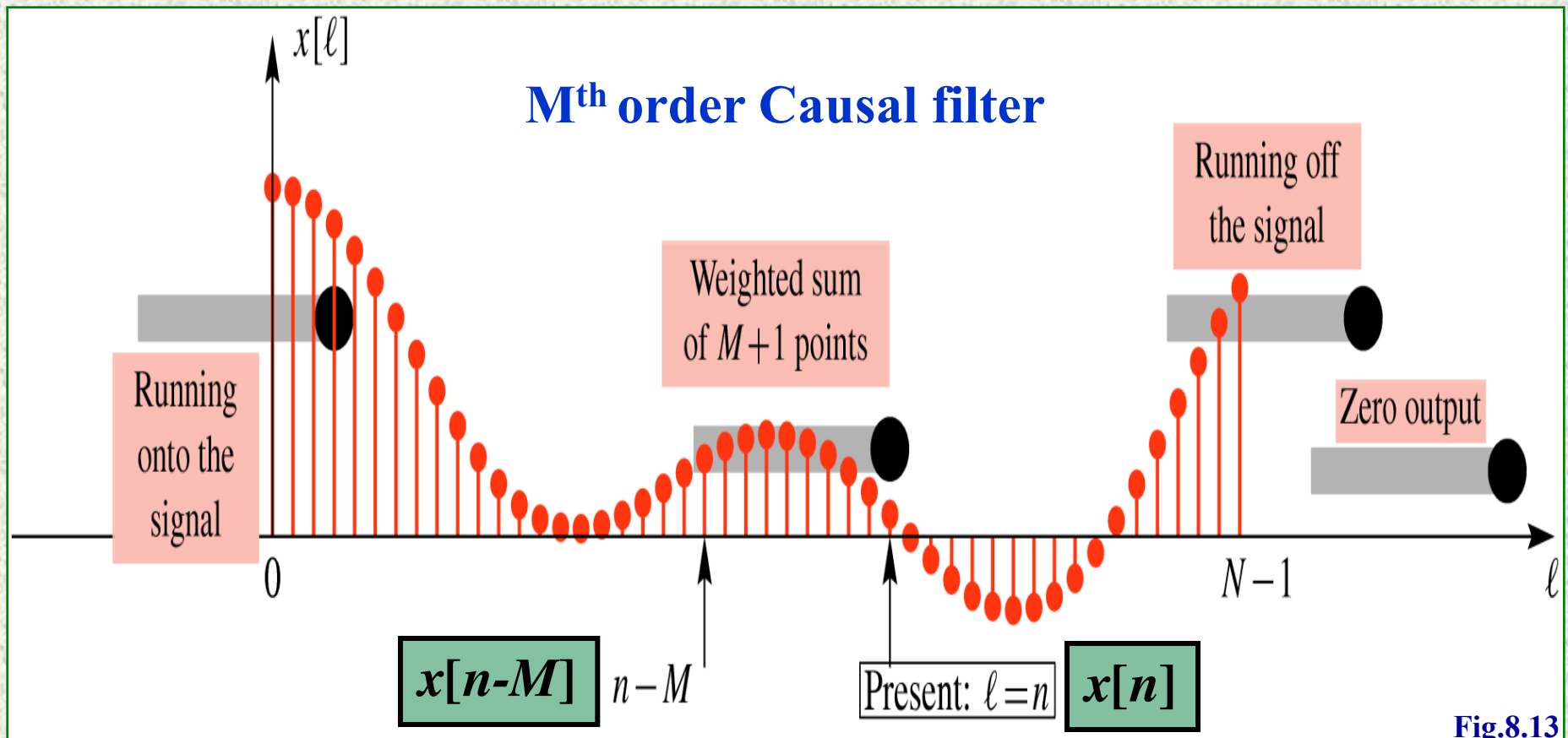
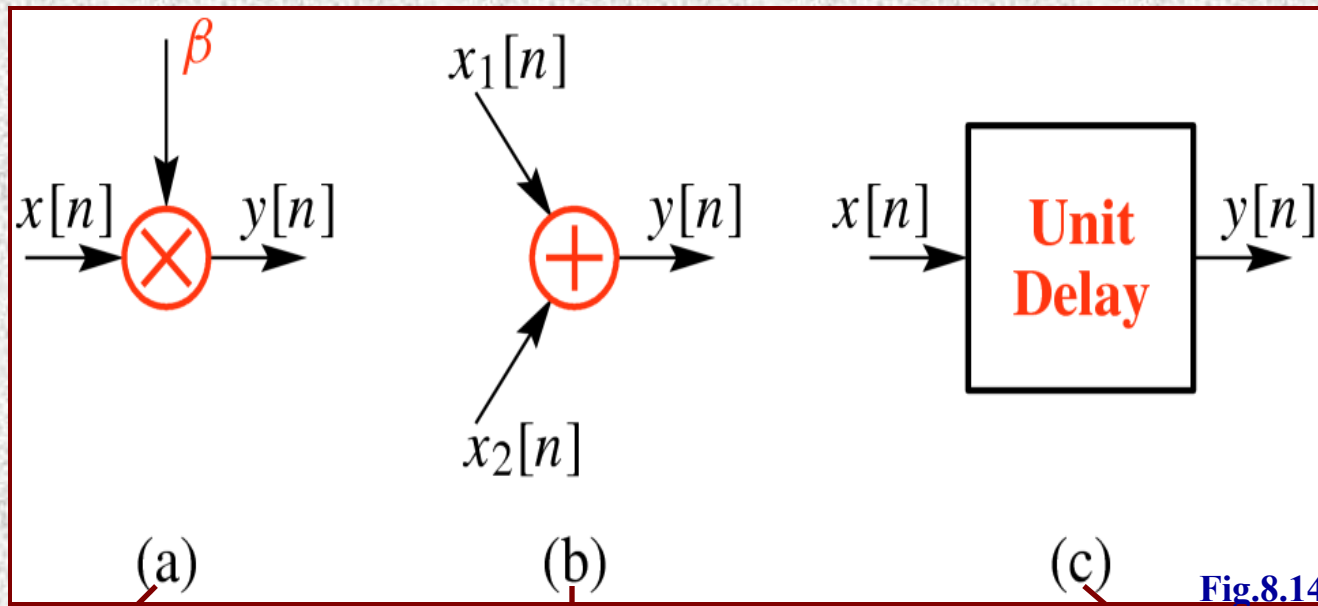


Fig.8.13

Block Diagrams: An Implementation view of FIR filters

Building blocks required



$$y[n] = \beta x[n]$$

$$y[n] = x_1[n] + x_2[n]$$

$$y[n] = x[n-1]$$

*direct form of
block diagram*

**Difference
equation**

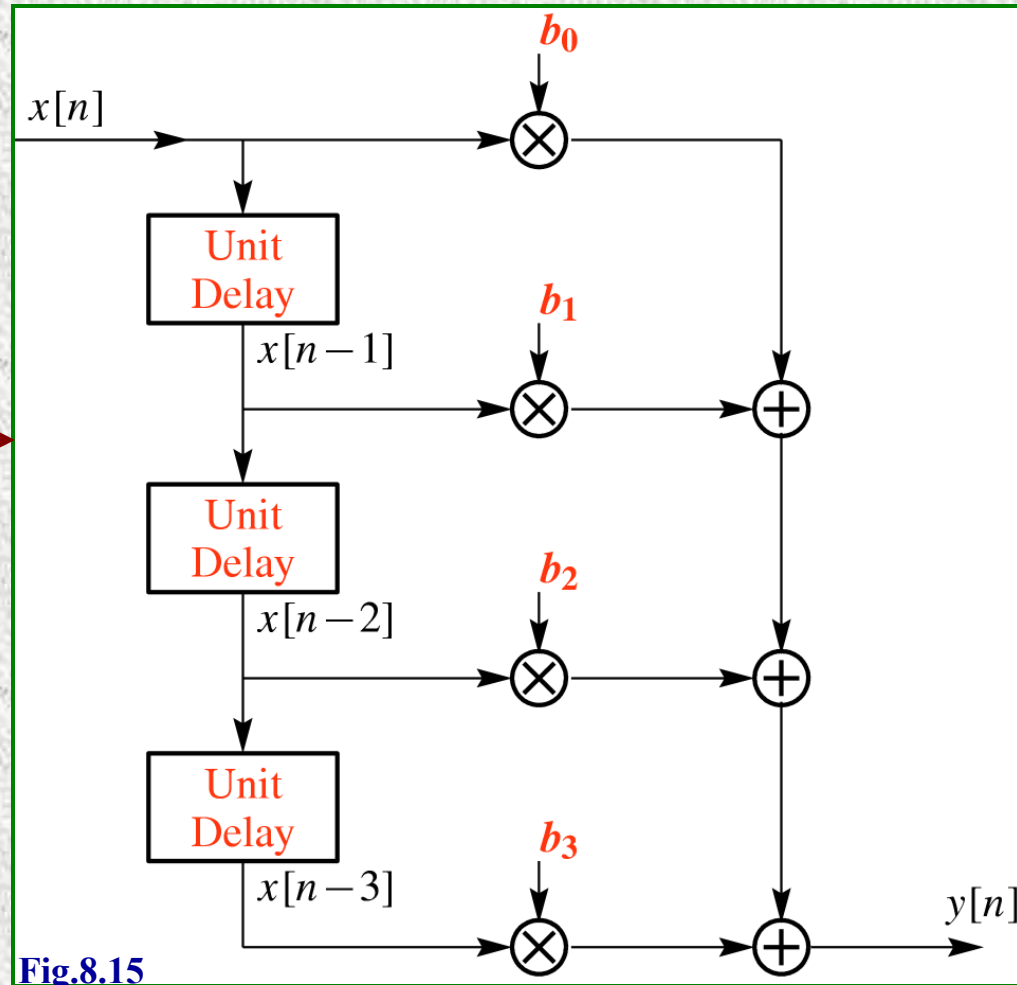


Fig.8.15

$$y[n] = \left(\left(\left(b_0 x[n] + b_1 x[n-1] \right) + b_2 x[n-2] \right) + b_3 x[n-3] \right)$$

transposed form
of block diagram

$$y[n] = b_0x[n] + v_1[n-1]$$

$$v_1[n] = b_1x[n] + v_2[n-1]$$

$$v_2[n] = b_2x[n] + v_3[n-1]$$

$$v_3[n] = b_3x[n]$$

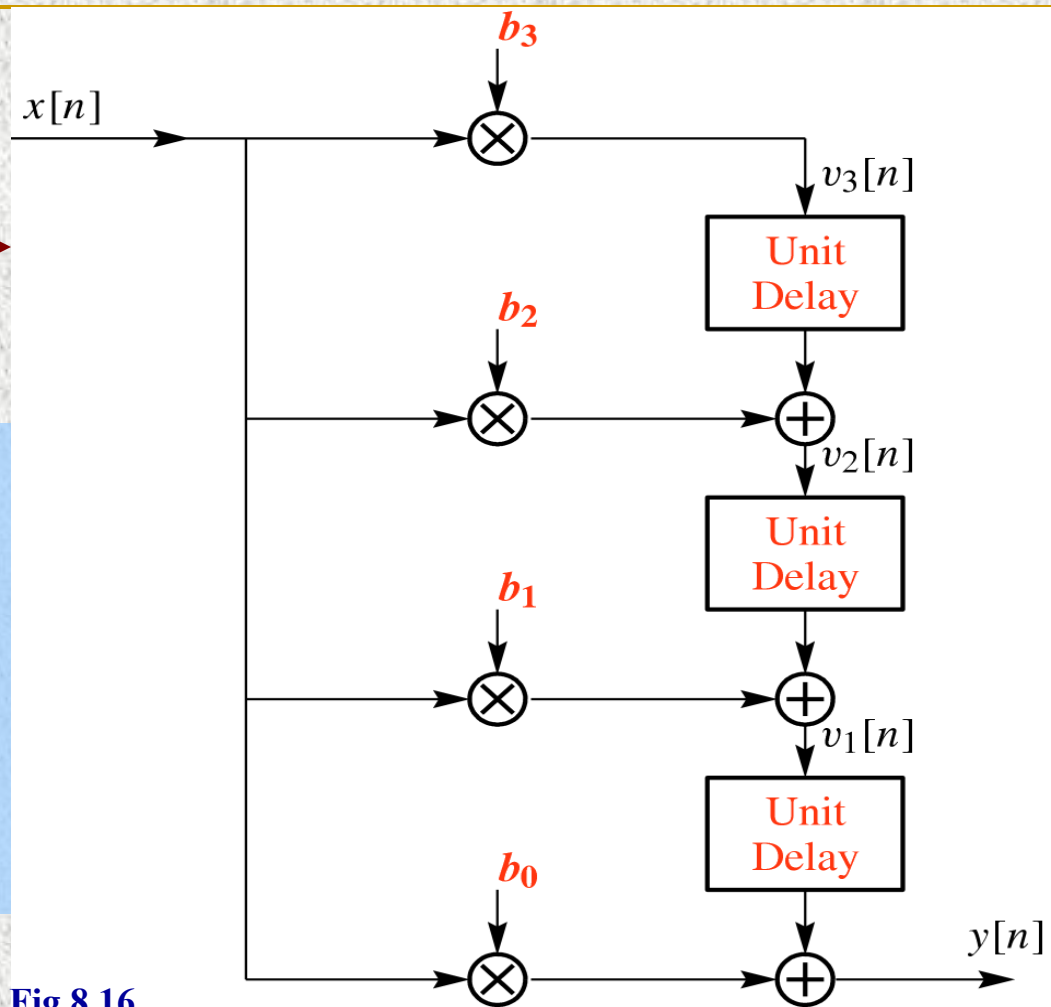


Fig.8.16

$$y[n] = \left(\left(\left(b_0x[n] + b_1x[n-1] \right) + b_2x[n-2] \right) + b_3x[n-3] \right)$$

Same Result!!!

Example: FIR filter application

Averaging of a sequence with different filter lengths

$$x[n] = (1.02)^n + \cos(2\pi n/8 + \pi/4) \quad \text{for } 0 \leq n \leq 40$$

Signal

Noise, cosine part

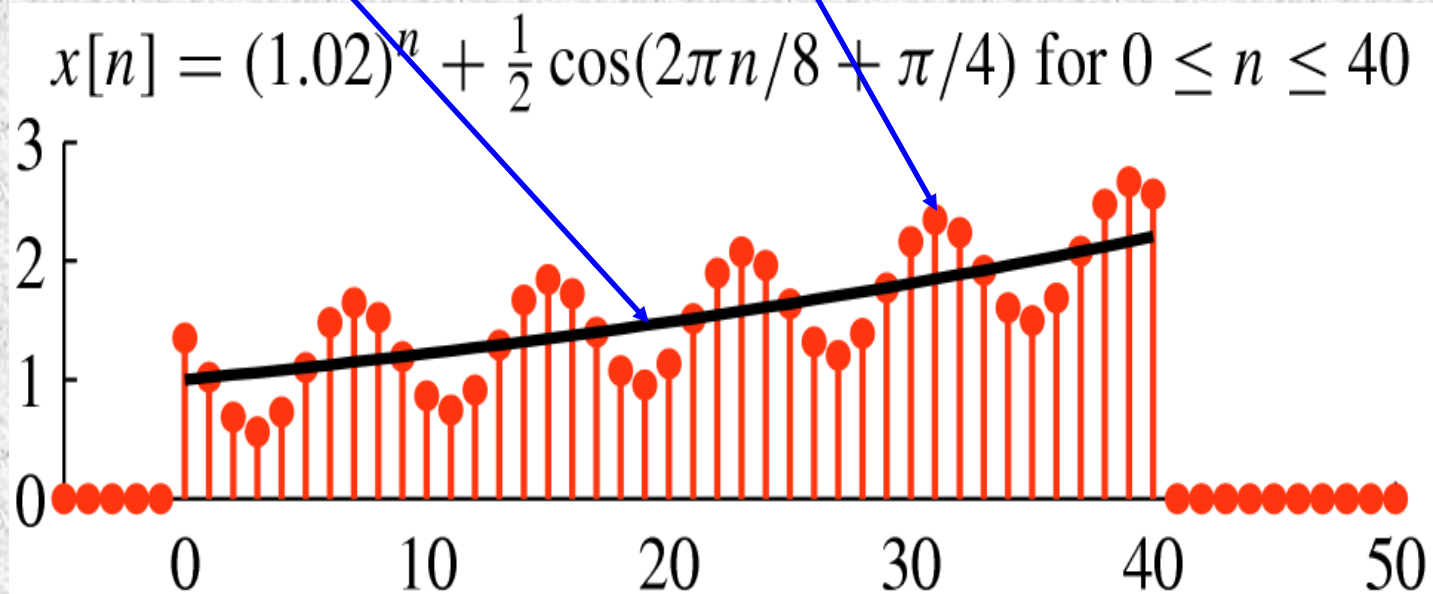


Fig.8.17

$$y_3[n] = \sum_{k=0}^2 \left(\frac{1}{3}\right)x[n-k]$$

1st 2 samples

Output goes till n=42

Output of 3-Point Running-Average Filter

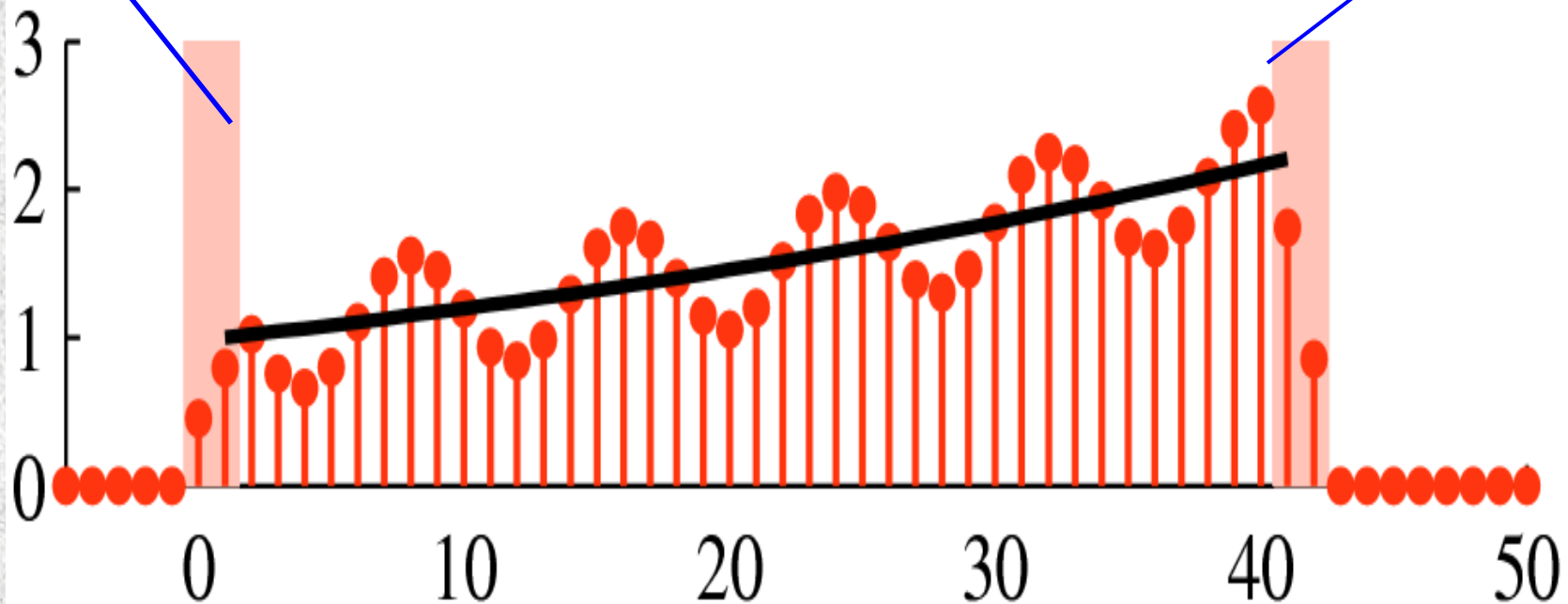


Fig.8.18

Notice that the output is smoother or its noise level is low than input

$$y_7[n] = \sum_{k=0}^6 \left(\frac{1}{7}\right)x[n-k]$$

1st 6 samples

Output goes till n=46

Output of 7-Point Running-Average Filter

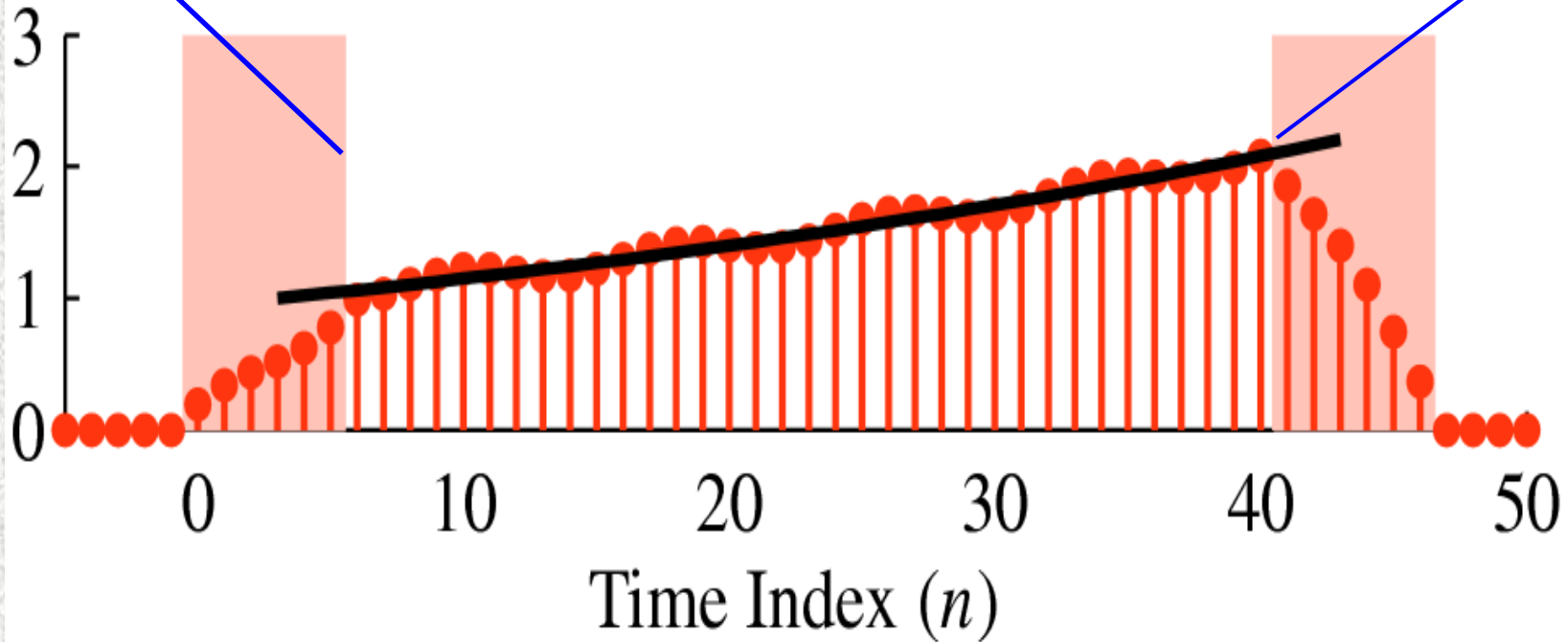


Fig.8.19

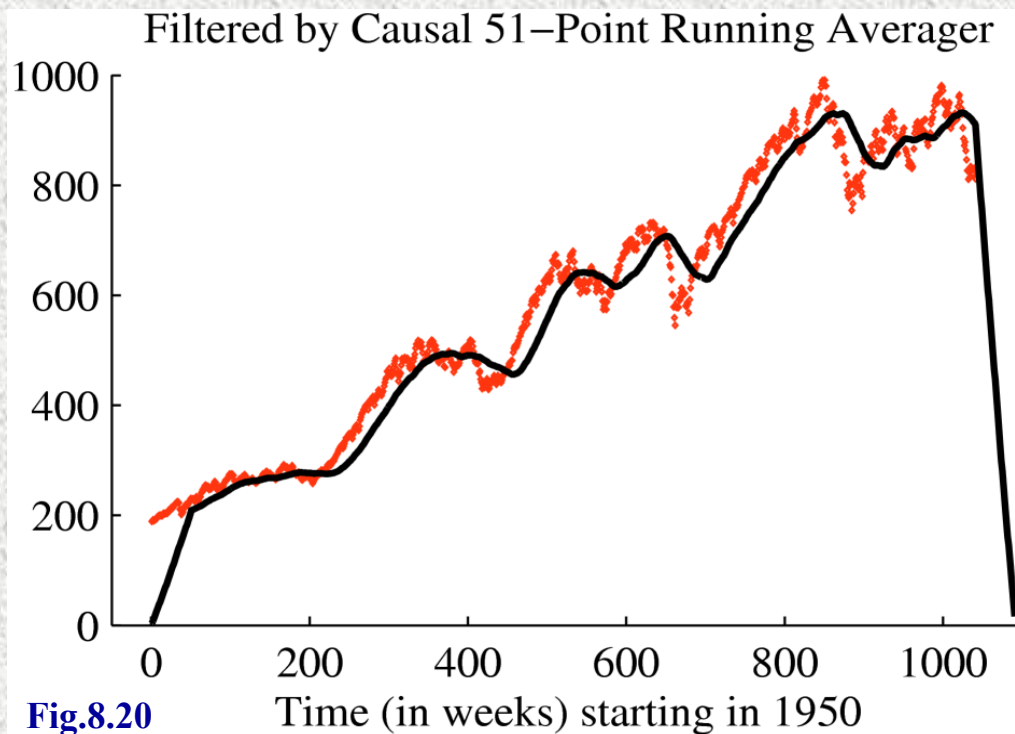
Notice that the output is much smoother than 3-point averaging method, noise level is low



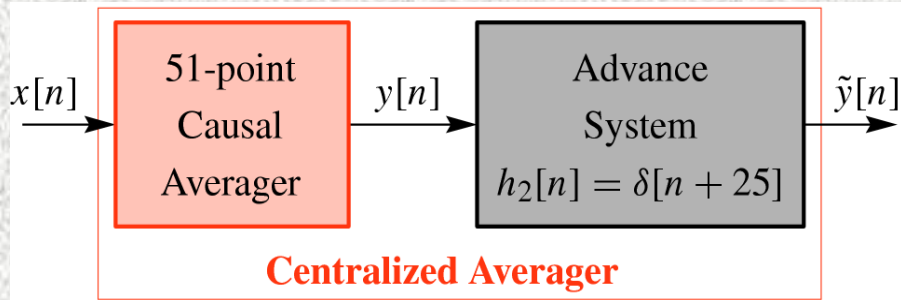
Example: DJIA signal

similar approach, averaging

$$y[n] = \left(\frac{1}{51}\right) \sum_{k=0}^{50} x[n-k]$$



Compensating for delay,



$$\tilde{y}[n] = \left(\frac{1}{51}\right) \sum_{k=-25}^{25} x[n-k]$$

Fig.8.21

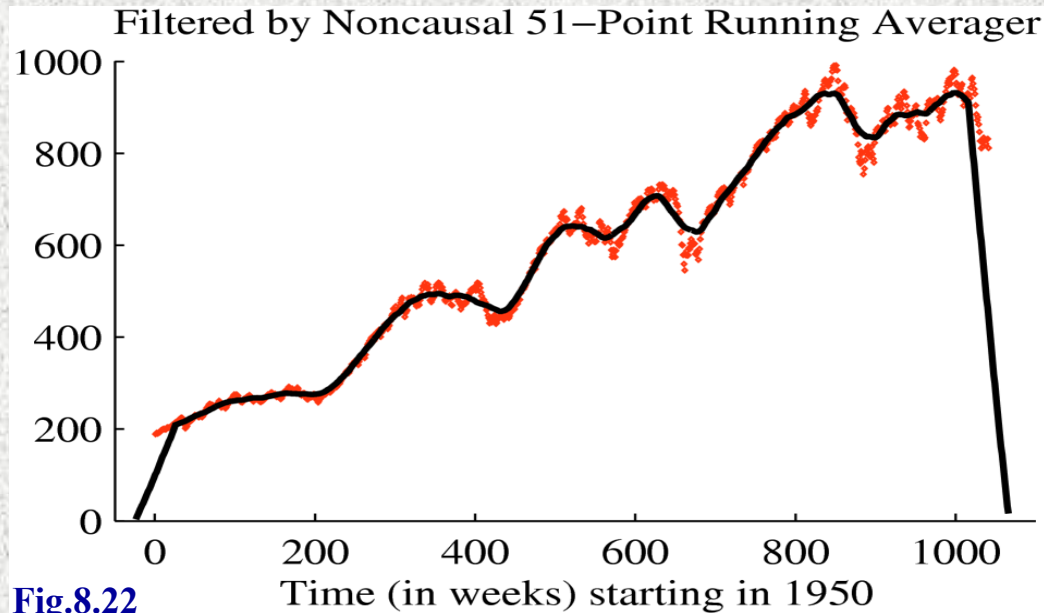


Fig.8.22

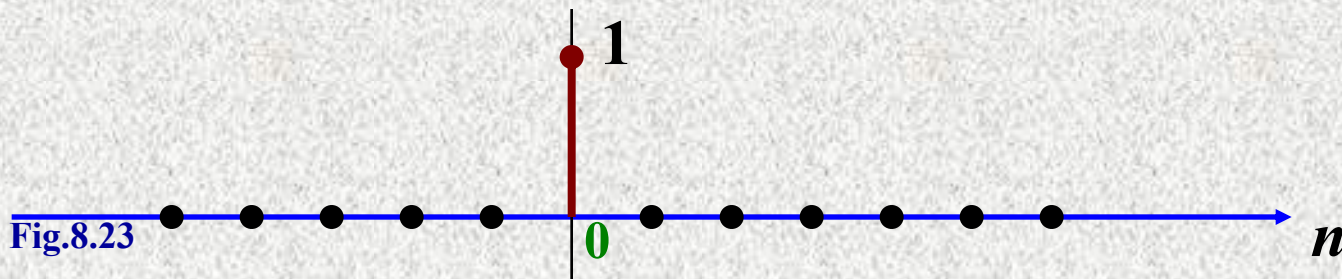
In these two examples, FIR filters are shown to remove rapid fluctuations



Discrete-Time Unit Impulse Sequence

Unit Impulse is the simplest sequence with only one nonzero value at $n=0$

$$\delta[n] = \begin{cases} 1 & n = 0 \\ 0 & n \neq 0 \end{cases}$$



$\delta[n]$ Is known as Kronecker delta function

Tabular form and a shifted version of unit impulse

n	...	-2	-1	0	1	2	3	4	5	6	...
$\delta[n]$	0	0	0	1	0	0	0	0	0	0	0
$\delta[n - 3]$	0	0	0	0	0	0	1	0	0	0	0

$\delta[n]$ is NON-ZERO
When its argument
is equal to ZERO

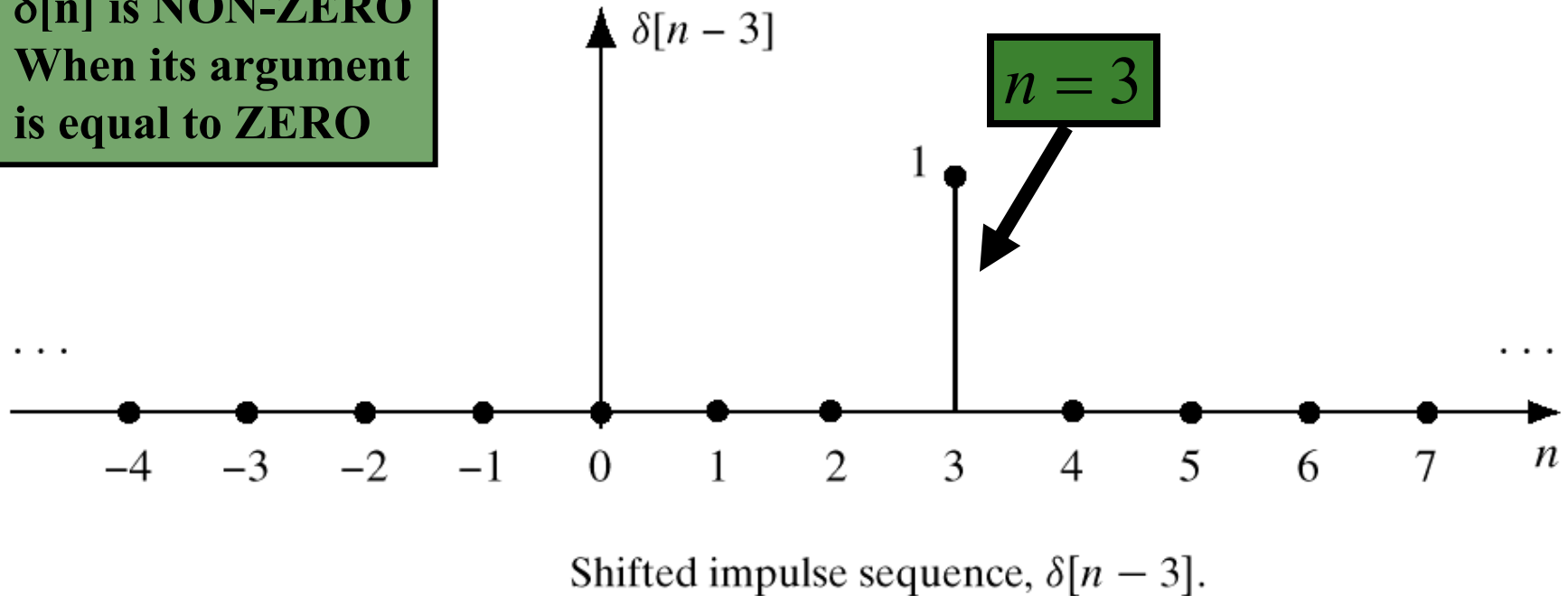
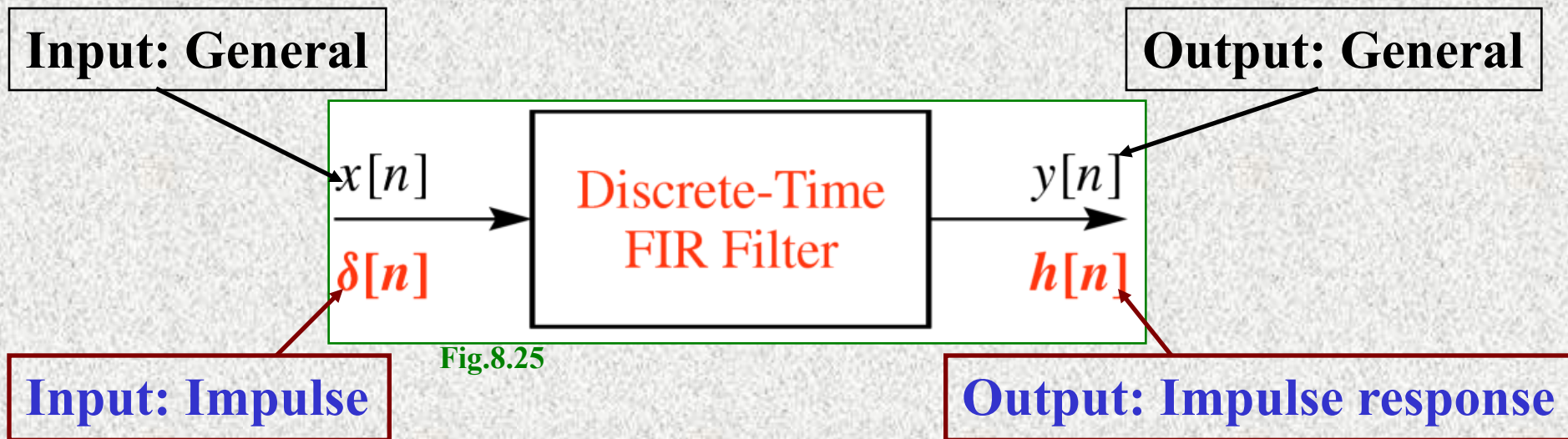


Fig.8.24



Unit Impulse Response Sequence

The response of an FIR filter to a unit impulse sequence is called as unit impulse response or simply '*impulse response*'



General FIR equation

$$y[n] = \sum_{k=0}^M b_k x[n-k]$$

Impulse Response

$$x[n] = \delta[n], \quad y[n] = h[n]$$

$$h[n] = \sum_{k=0}^M b_k \delta[n-k] = \begin{cases} b_n & n = 0, 1, 2, \dots, M \\ 0 & \textit{otherwise} \end{cases}$$

The sum evaluates to a single term for each value of n, as $\delta[n-k]$ is nonzero only when $n=k$

Tabular form for '*Impulse Response*' equation

n	$n < 0$	0	1	2	3	...	M	$M + 1$	$n > M + 1$
$x[n] = \delta[n]$	0	1	0	0	0	0	0	0	0
$y[n] = h[n]$	0	b_0	b_1	b_2	b_3	...	b_M	0	0

Fig.8.26

*In the above table $h[n] = 0$ for $n < 0$ and $n > M$,
The length of impulse response sequence is finite.
This is why the system is called a
Finite Impulse Response (FIR) system*

Example1: 3-point average filter

$$y_3[n] = \left(\frac{1}{3}\right) \sum_{k=0}^2 x[n-k] \text{ compare with standard equation}$$

$$y[n] = \sum_{k=0}^M b_k x[n-k], \quad h[0] = b_0, \quad h[1] = b_1, \quad h[2] = b_2$$

$$\therefore h[0] = \frac{1}{3}, \quad h[1] = \frac{1}{3}, \quad h[2] = \frac{1}{3}$$

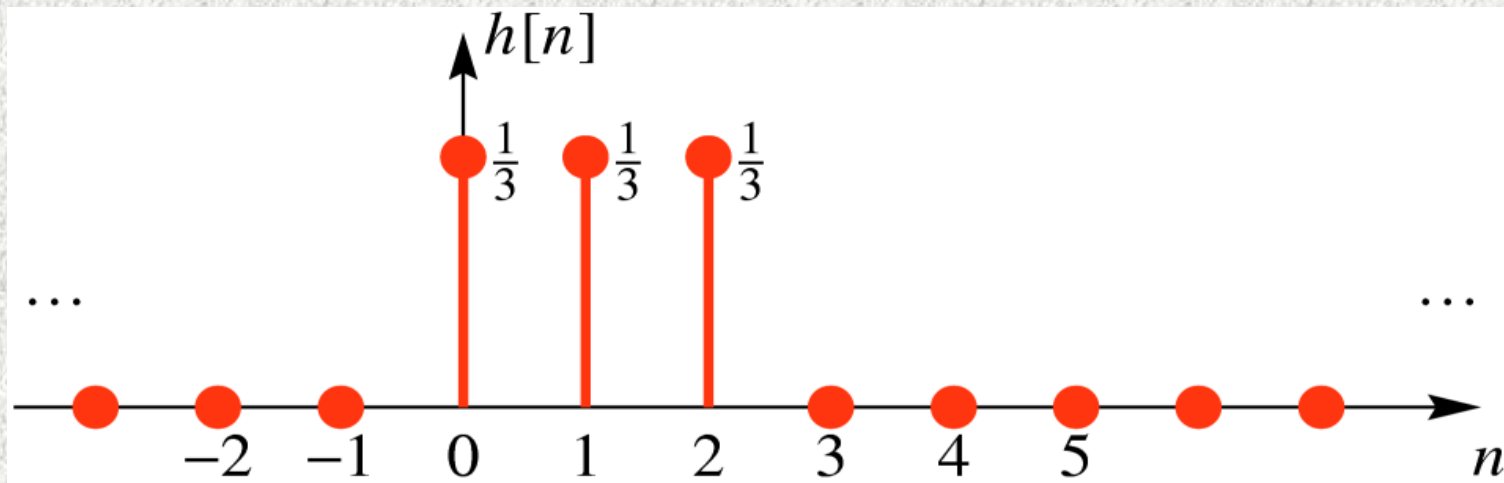


Fig.8.27

Example2

Find the difference equation governing the input – output relation with FIR filter coefficients $\{3, -1, 2, 1\}$

FIR filter coefficients $h[n]=\{3, -1, 2, 1\}$

$$h[n] = b_k \quad \text{for } k = 0, 1 \dots N$$

$$y[n] = \sum_{k=0}^M b_k x[n-k]$$

$$= \sum_{k=0}^3 b_k x[n-k]$$

$$= b_0 x[n] + b_1 x[n-1] + b_2 x[n-2] + b_3 x[n-3]$$

$$= 3x[n] - x[n-1] + 2x[n-2] + x[n-3] \text{ difference equation}$$

Representation of a general sequence $x[n]$

Any sequence can be obtained by adding shifted impulses

$$x[n] = 2\delta[n] + 4\delta[n - 1] + 6\delta[n - 2] + 4\delta[n - 3] + 2\delta[n - 4]$$

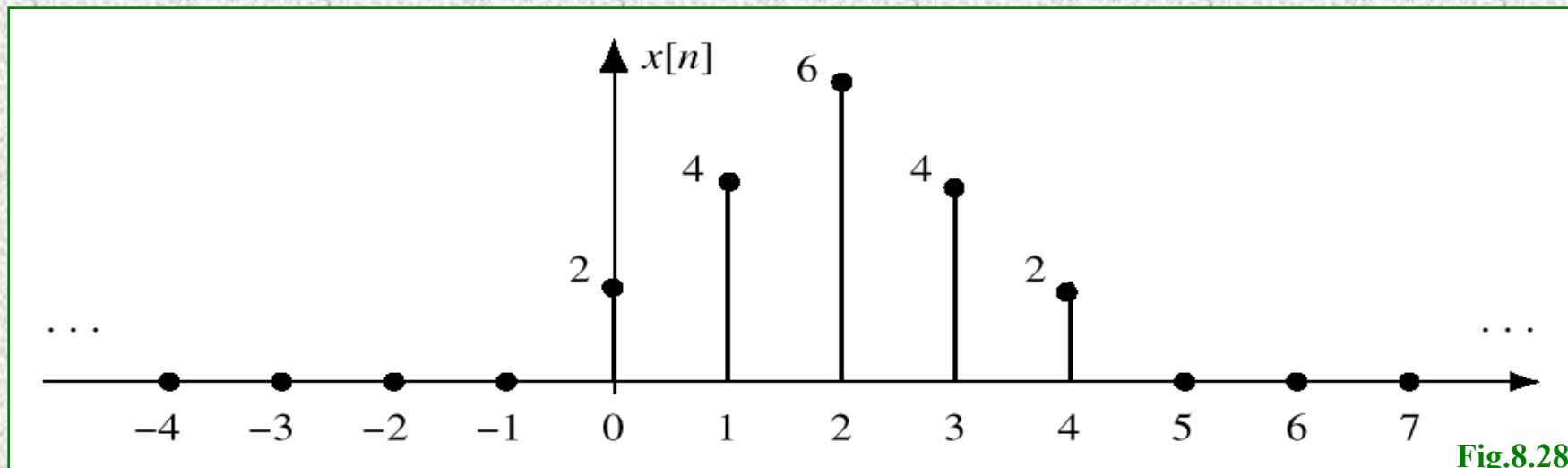


Fig.8.28

Tabular form: Breaking a sequence into shifted impulses

n	...	-2	-1	0	1	2	3	4	5	6	...
$2\delta[n]$	0	0	0	2	0	0	0	0	0	0	0
$4\delta[n - 1]$	0	0	0	0	4	0	0	0	0	0	0
$6\delta[n - 2]$	0	0	0	0	0	6	0	0	0	0	0
$4\delta[n - 3]$	0	0	0	0	0	0	4	0	0	0	0
$2\delta[n - 4]$	0	0	0	0	0	0	0	2	0	0	0
$x[n]$	0	0	0	2	4	6	4	2	0	0	0

$$x[n] = \sum_k x[k]\delta[n - k]$$

For any signal

$$= \dots + x[-1]\delta[n + 1] + x[0]\delta[n] + x[1]\delta[n - 1] + \dots$$

Fig.8.29

Discrete-Time Convolution Sum

General Discrete-System

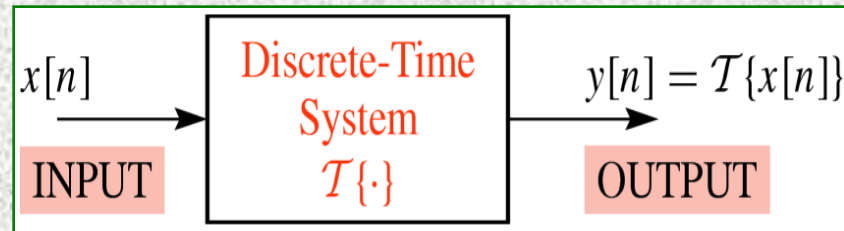
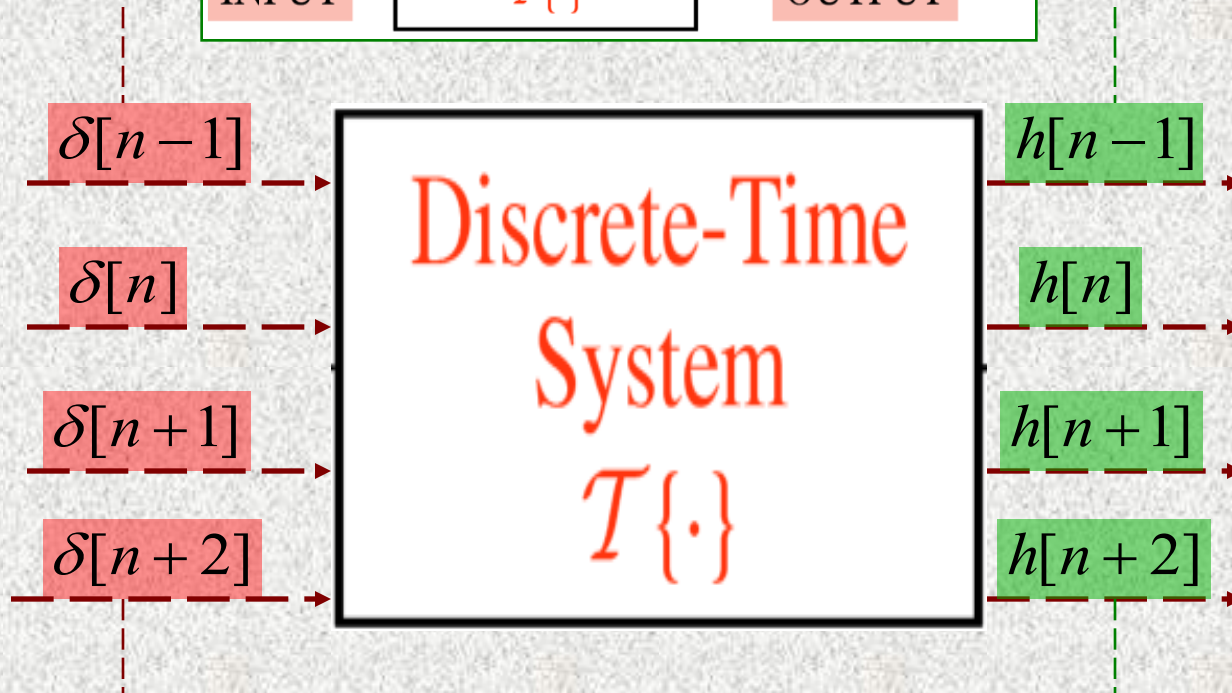


Fig.8.30



From the previous figure,

$$x[0]\delta[n] = x[0]h[n]$$

$$x[0]\delta[n-1] = x[0]h[n-1]$$

•

$$x[0]\delta[n-k] = x[0]h[n-k]$$

As shown previously using superposition,

$$x[n] = \sum_{k=-\infty}^{\infty} x[k]\delta[n-k]$$

$$y[n] = \sum_{k=-\infty}^{\infty} x[k]h[n-k], \quad \text{Convolution Sum}$$



Example1: FIR from Convolution

$$y[n] = \sum_{k=-\infty}^{\infty} x[k]h[n-k], \text{ If } h[n] \text{ is non-zero}$$

only in the interval $0 \leq n \leq M$ then,

$$y[n] = \sum_{k=n-M}^n x[k]h[n-k],$$

which is a classic FIR filter

About limits : $0 \leq (n-k) \leq M$

$$\therefore (n-M) \leq k \leq n$$



Example2: Computing the output

$$x[n] = \{2, 4, 6, 4, 2\}, \quad h[n] = \{3, -1, 2, 1\}$$

convolve $x[n]$ and $h[n]$ to get $y[n]$

- Write out the signals $x[n]$ and $y[n]$ on separate rows
- The output is to be computed as sum of shifted rows
- Each shifted row is to be produced by multiplying the $x[n]$ row by one of the $h[k]$ values and,
- By shifting the result to the right so that it lines up with $h[k]$ position

Numerical convolution done through the above process is also called as synthetic polynomial multiplication

Tabular form describing the convolution

n	$n < 0$	0	1	2	3	4	5	6	7	$n > 7$
$x[n]$	0	2	4	6	4	2	0	0	0	0
$h[n]$	0	3	-1	2	1	0	0	0	0	0
$h[0]x[n]$	0	6	12	18	12	6	0	0	0	0
$h[1]x[n-1]$	0	0	-2	-4	-6	-4	-2	0	0	0
$h[2]x[n-2]$	0	0	0	4	8	12	8	4	0	0
$h[3]x[n-3]$	0	0	0	0	2	4	6	4	2	0
$y[n]$	0	6	10	18	16	18	12	8	2	0

$y[2]$

Fig.8.32

Try the demo on your CD

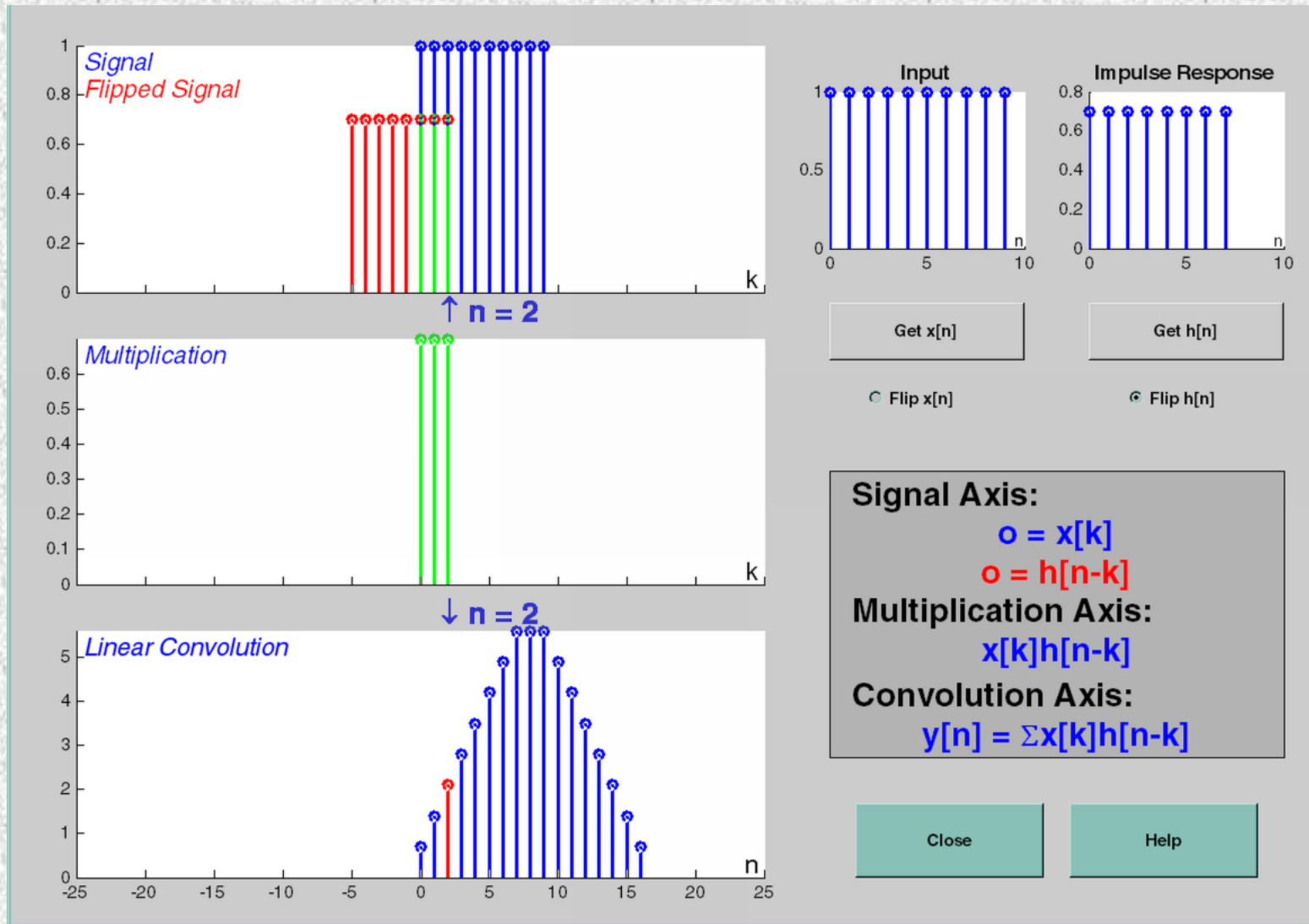


Fig.8.33

Reference

James H. McClellan, Ronald W. Schafer and Mark A. Yoder, “ 5.1-5.4 and 5.6--Signal Processing First”, Prentice Hall, 2003
