

# Signal Processing First



## Lecture 10 Filtering Intro

# READING ASSIGNMENTS



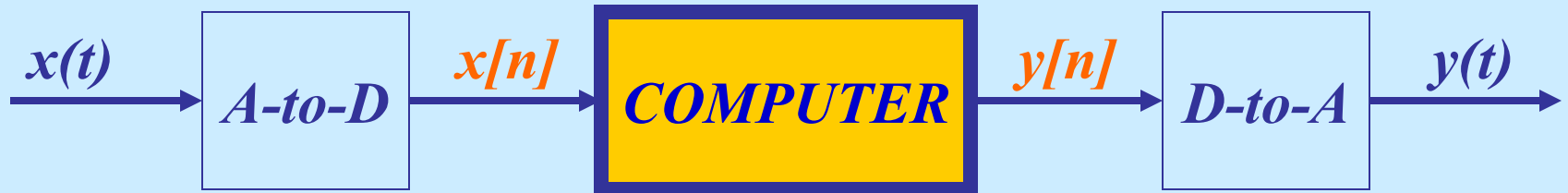
- This Lecture:
  - Chapter 5, Sects. 5-1, 5-2 and 5-3 (partial)
- Other Reading:
  - Recitation: Ch. 5, Sects 5-4, 5-6, 5-7 and 5-8
    - CONVOLUTION
  - Next Lecture: Ch 5, Sects. 5-3, 5-5 and 5-6

# LECTURE OBJECTIVES



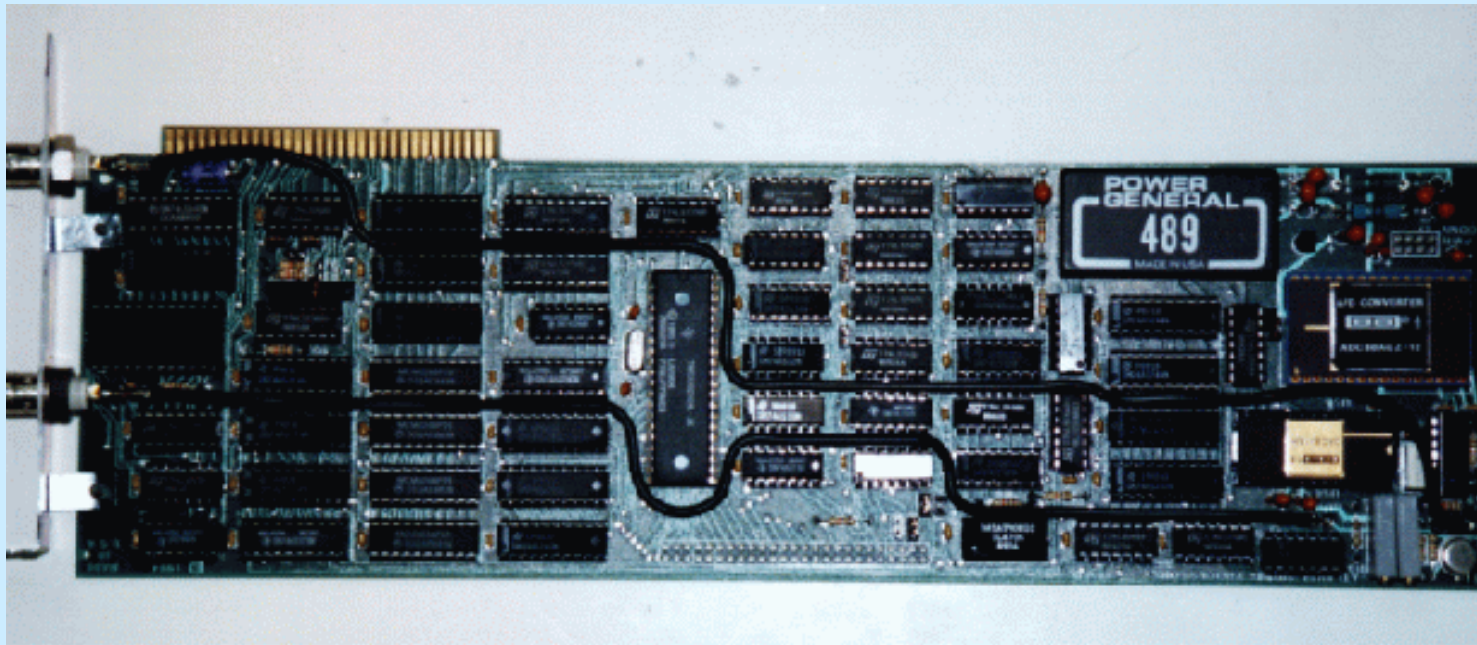
- INTRODUCE FILTERING IDEA
  - **Weighted** Average
  - **Running** Average
- FINITE IMPULSE RESPONSE FILTERS
  - **FIR** Filters
  - Show how to **compute** the output  $y[n]$  from the input signal,  $x[n]$

# DIGITAL FILTERING



- CONCENTRATE on the COMPUTER
  - PROCESSING ALGORITHMS
  - SOFTWARE (MATLAB)
  - HARDWARE: DSP chips, VLSI
- **DSP**: DIGITAL SIGNAL PROCESSING

# The TMS32010, 1983



First PC plug-in board from Atlanta Signal Processors Inc.

# Rockland Digital Filter, 1971



The image shows a photograph of the Rockland Model 4136 Programmable Digital Filter. The device is a rectangular metal box with a front panel. On the front panel, there are three large rotary knobs labeled 'FILTER ORDER', 'NO. COEFFICIENTS', and 'OUTPUT GAIN'. Below these knobs are three pairs of indicator lights labeled 'ON POWER', 'OVERFLOW', and 'DEVELOP'. The top of the device has a black label that reads 'ROCKLAND PROGRAMMABLE DIGITAL FILTER - MODEL 4136'. To the right of the device, the text 'Model 4136 PROGRAMMABLE DIGITAL FILTER' is printed in large, bold, blue letters.

## 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

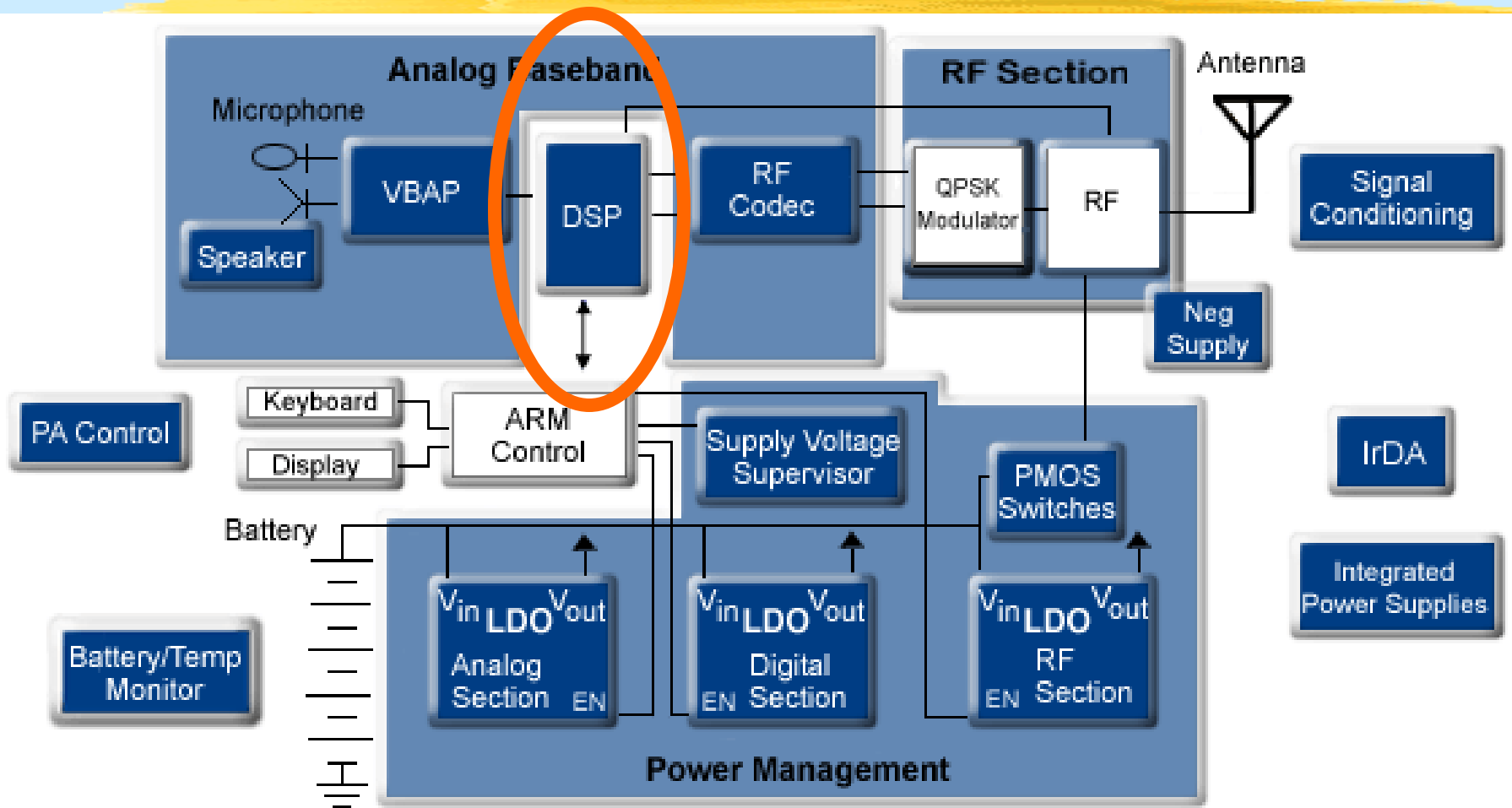
The transfer function from filter input to filter output in z-transform notation is given by

$$H_N(z) = \prod_{n=1}^N \frac{K_n(1+z^{-1}A1_n+z^{-2}A2_n)}{1-z^{-1}B1_n-z^{-2}B2_n} \quad (1)$$

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

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

# Digital Cell Phone (ca. 2000)



Now it plays video

# DISCRETE-TIME SYSTEM



- OPERATE on  $x[n]$  to get  $y[n]$
- WANT a **GENERAL** CLASS of SYSTEMS
  - **ANALYZE** the SYSTEM
    - TOOLS: TIME-DOMAIN & FREQUENCY-DOMAIN
  - **SYNTHESIZE** the SYSTEM



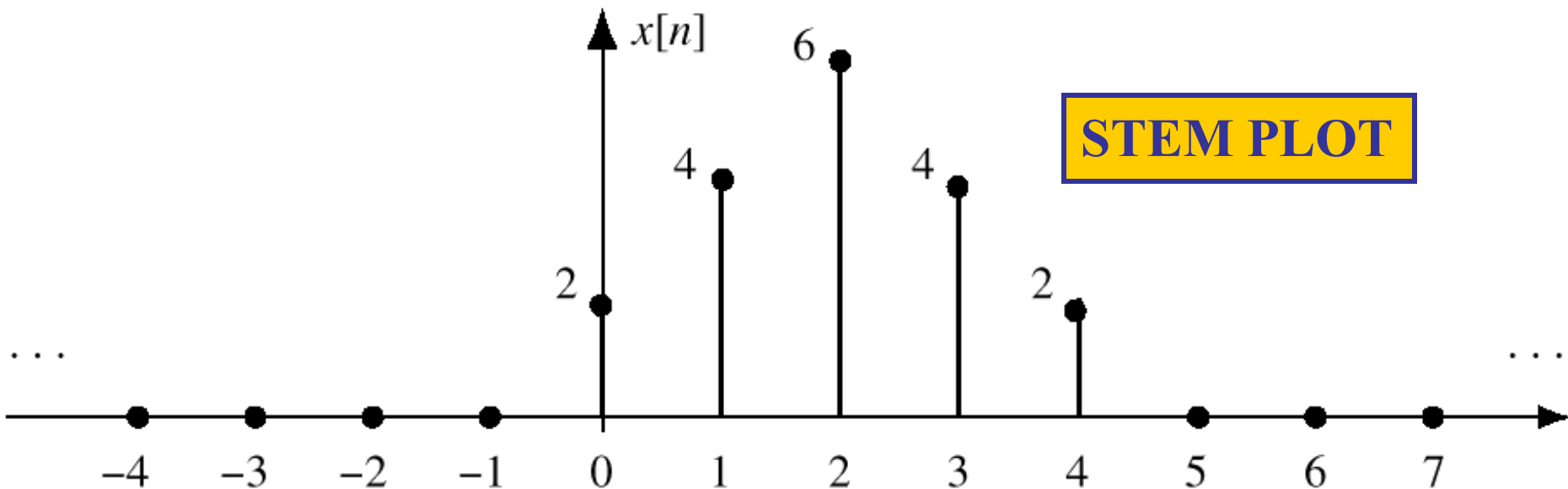
# D-T SYSTEM EXAMPLES



- EXAMPLES:
  - POINTWISE OPERATORS
    - SQUARING:  $y[n] = (x[n])^2$
  - RUNNING AVERAGE
    - **RULE:** “the output at time  $n$  is the average of three consecutive input values”

# DISCRETE-TIME SIGNAL

- $x[n]$  is a LIST of NUMBERS
  - INDEXED by “ $n$ ”



# 3-PT AVERAGE SYSTEM

- ADD 3 CONSECUTIVE NUMBERS
  - Do this for each “ $n$ ”

the following input–output equation

**Make a TABLE**

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

$n$	$n < -2$	$-2$	$-1$	$0$	$1$	$2$	$3$	$4$	$5$	$n > 5$
$x[n]$	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

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

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

## INPUT SIGNAL

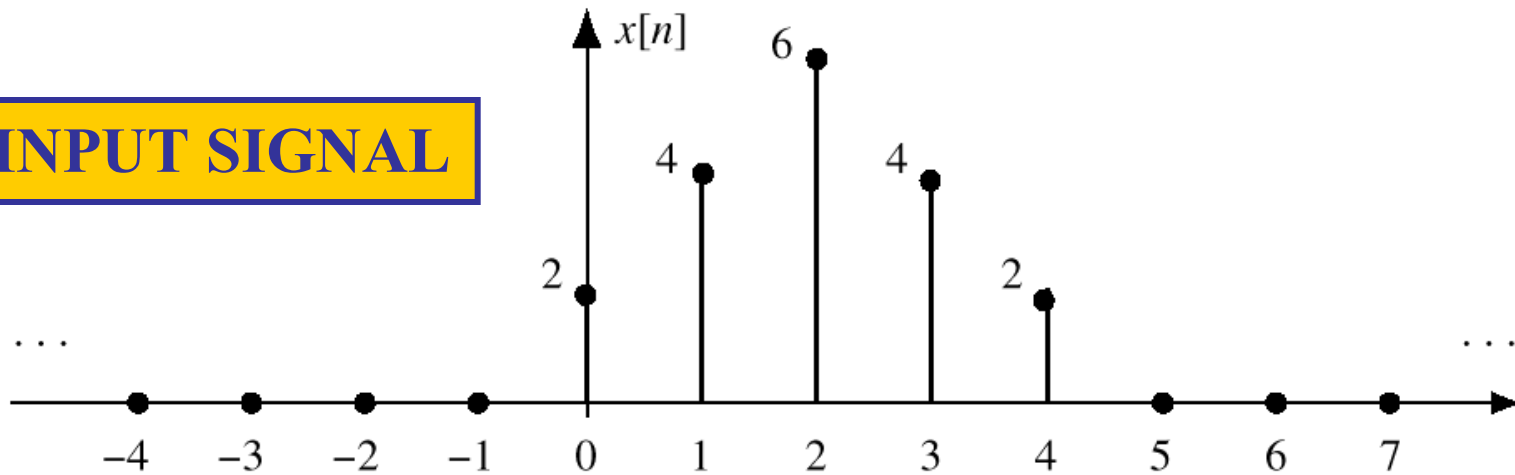
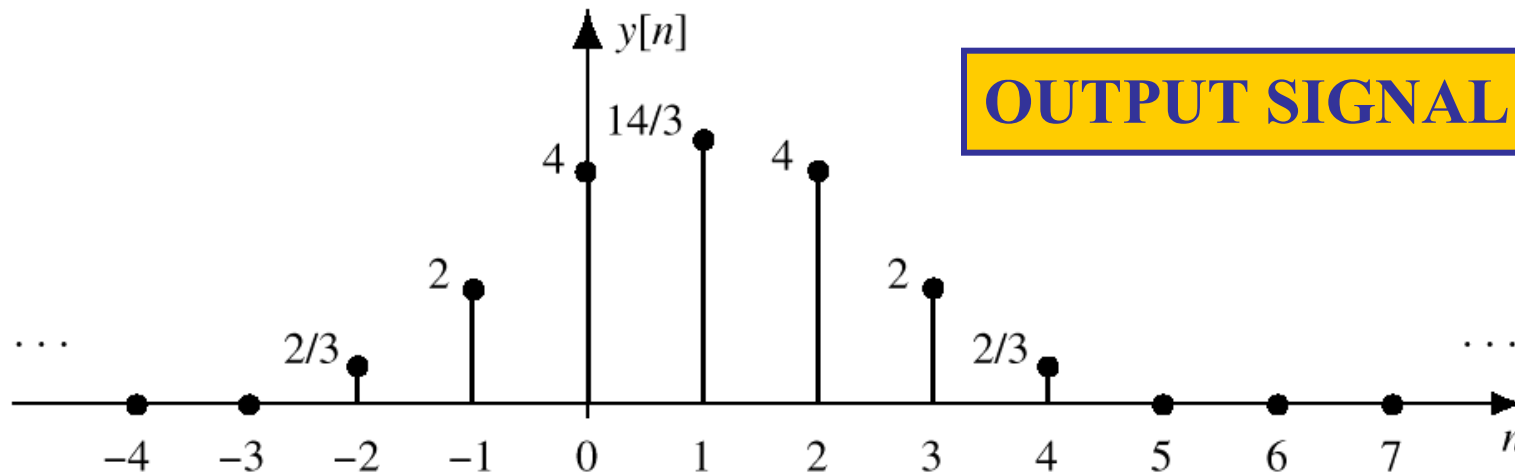


Figure 5.2 Finite-length input signal,  $x[n]$ .

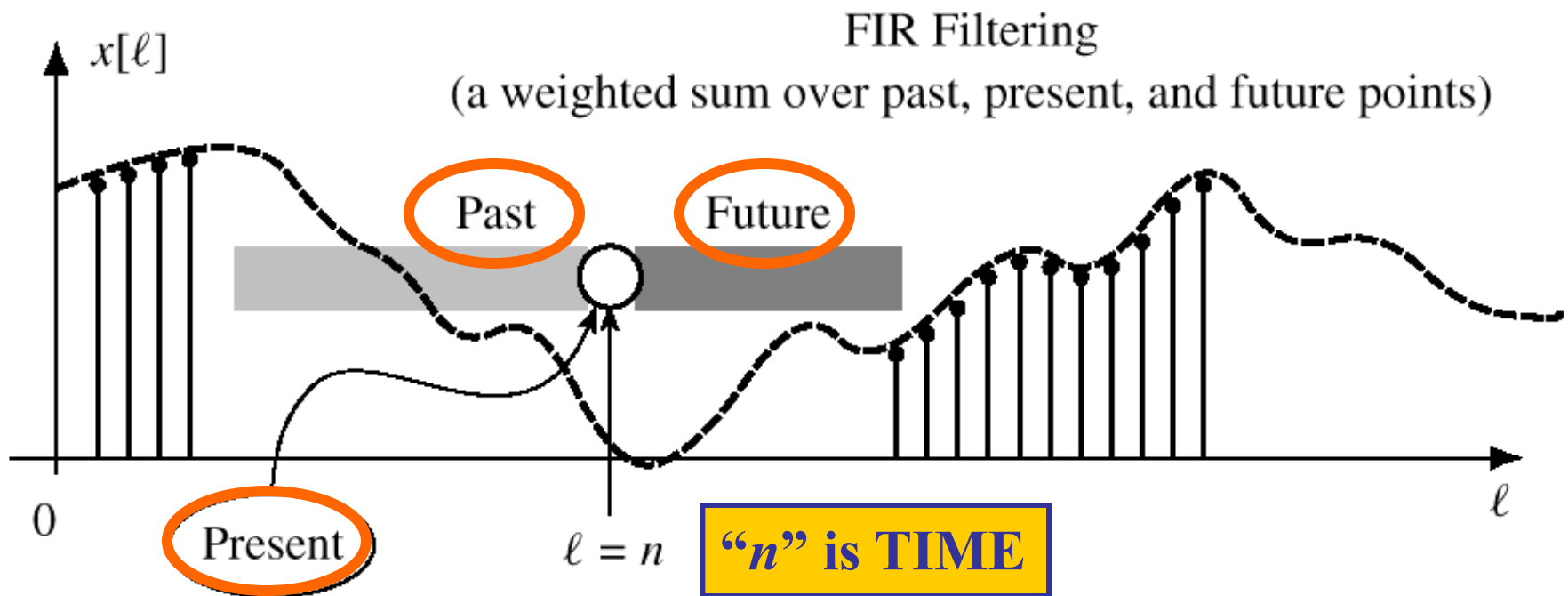
$$y[n] = \frac{1}{3}(x[n] + x[n+1] + x[n+2])$$



## OUTPUT SIGNAL

Figure 5.3 Output of running average,  $y[n]$ .

# PAST, PRESENT, FUTURE



**Figure 5.4** The running-average filter calculation at time index  $n$  uses values within a sliding window (shaded). Dark shading indicates the future ( $l > n$ ); light shading, the past ( $l < n$ ).

# ANOTHER 3-pt AVERAGER

- Uses “PAST” VALUES of  $x[n]$ 
  - IMPORTANT IF “ $n$ ” represents **REAL TIME**
    - WHEN  $x[n]$  &  $y[n]$  ARE STREAMS

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

$n$	$n < -2$	-2	-1	0	1	2	3	4	5	6	7	$n > 7$
$x[n]$	0	0	0	2	4	<b>6</b>	<b>4</b>	<b>2</b>	0	0	0	0
$y[n]$	0	0	0	$\frac{2}{3}$	2	4	$\frac{14}{3}$	<b>4</b>	2	$\frac{2}{3}$	0	0

# GENERAL FIR FILTER

- FILTER COEFFICIENTS  $\{b_k\}$

- DEFINE THE FILTER

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

- For example,  $b_k = \{3, -1, 2, 1\}$

$$\begin{aligned} y[n] &= \sum_{k=0}^3 b_k x[n-k] \\ &= 3x[n] - x[n-1] + 2x[n-2] + x[n-3] \end{aligned}$$

# GENERAL FIR FILTER

- FILTER COEFFICIENTS  $\{b_k\}$

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

- FILTER ORDER is M
- FILTER LENGTH is  $L = M + 1$ 
  - NUMBER of FILTER COEFFS is L

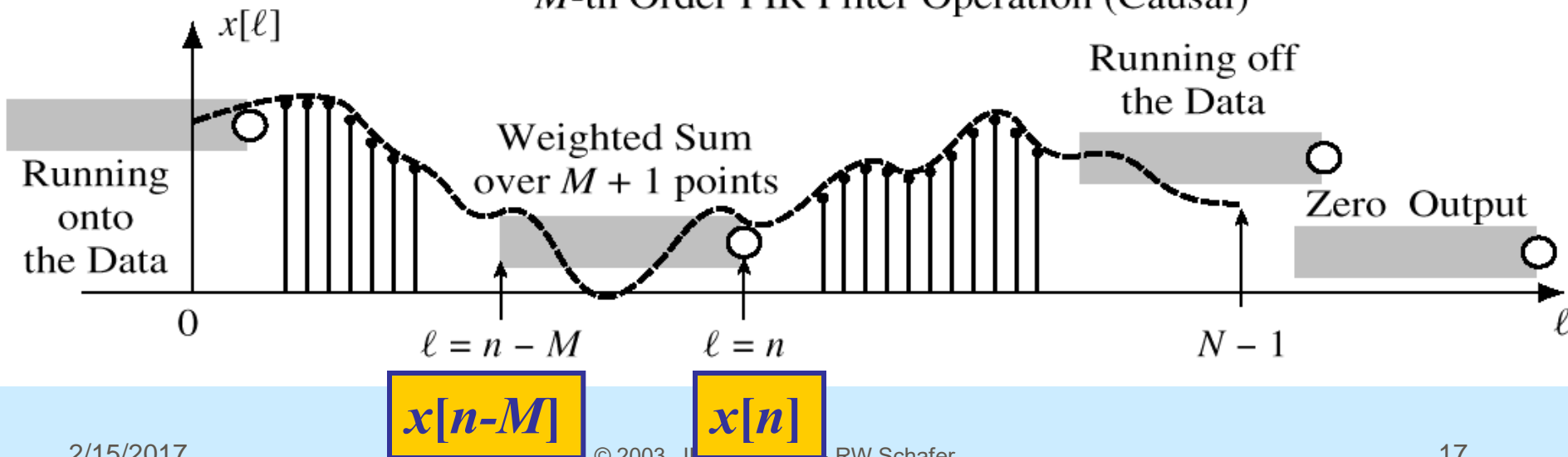


# GENERAL FIR FILTER

- SLIDE a WINDOW across  $x[n]$

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

*M*-th Order FIR Filter Operation (Causal)



# FILTERED STOCK SIGNAL

Period: **YTD**

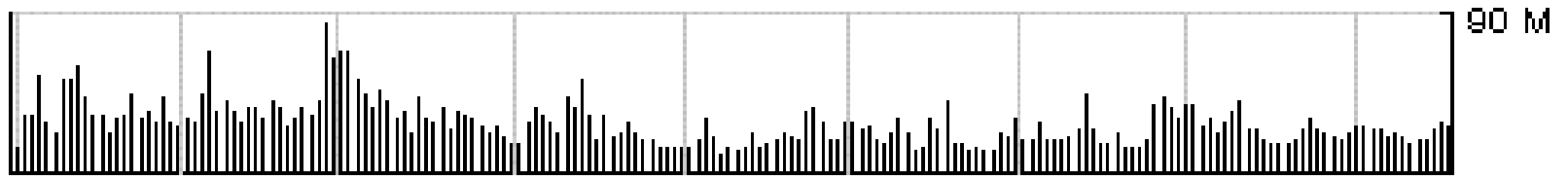
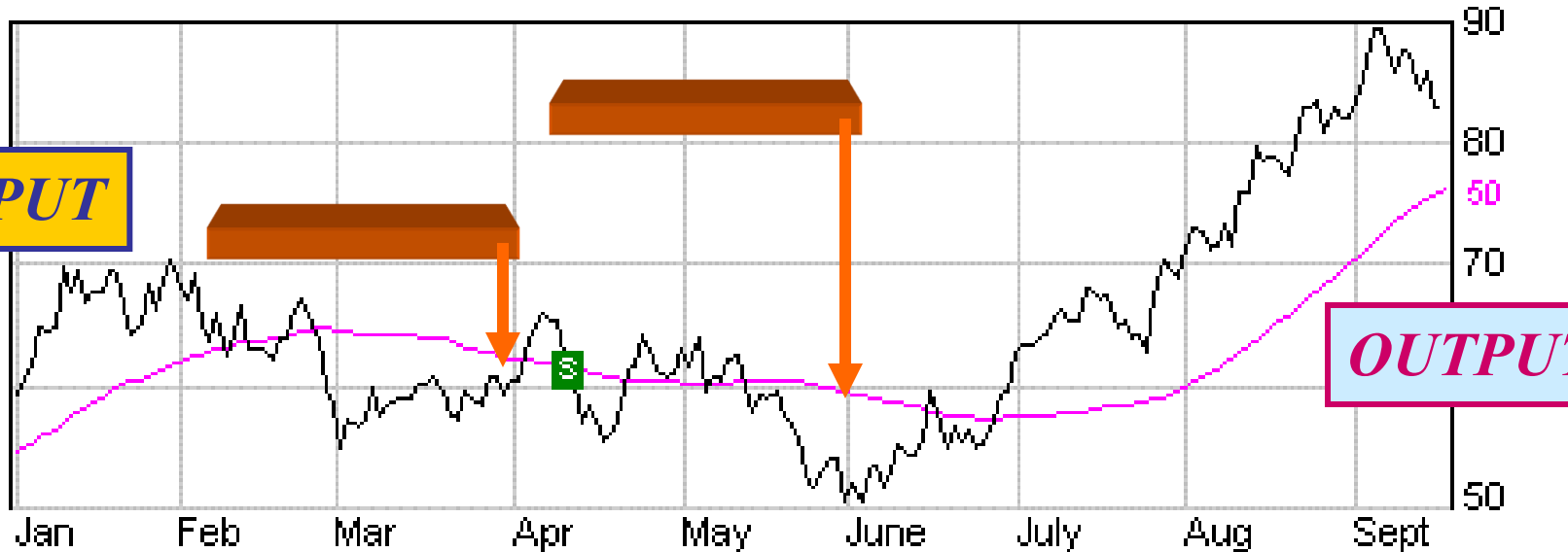
Chart Type: **Closing Prices**

**INTC** **B4 3/4** **+ 1/8**

[S] = Stock split

**INPUT**

**OUTPUT**



1999

**50-pt Averager**

Moving Averages:  None  25  50  100  200

# SPECIAL INPUT SIGNALS

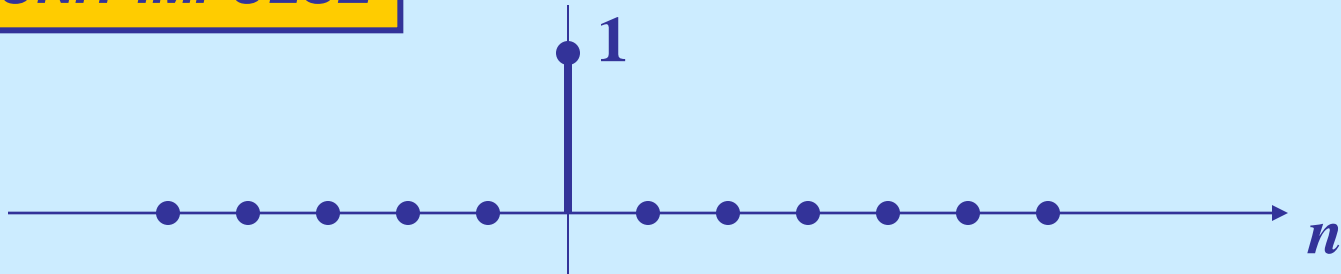
- $x[n] = \text{SINUSOID}$

*FREQUENCY RESPONSE (LATER)*

- $x[n]$  has only one NON-ZERO VALUE

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

*UNIT-IMPULSE*



# UNIT IMPULSE SIGNAL $\delta[n]$

$n$	...	-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

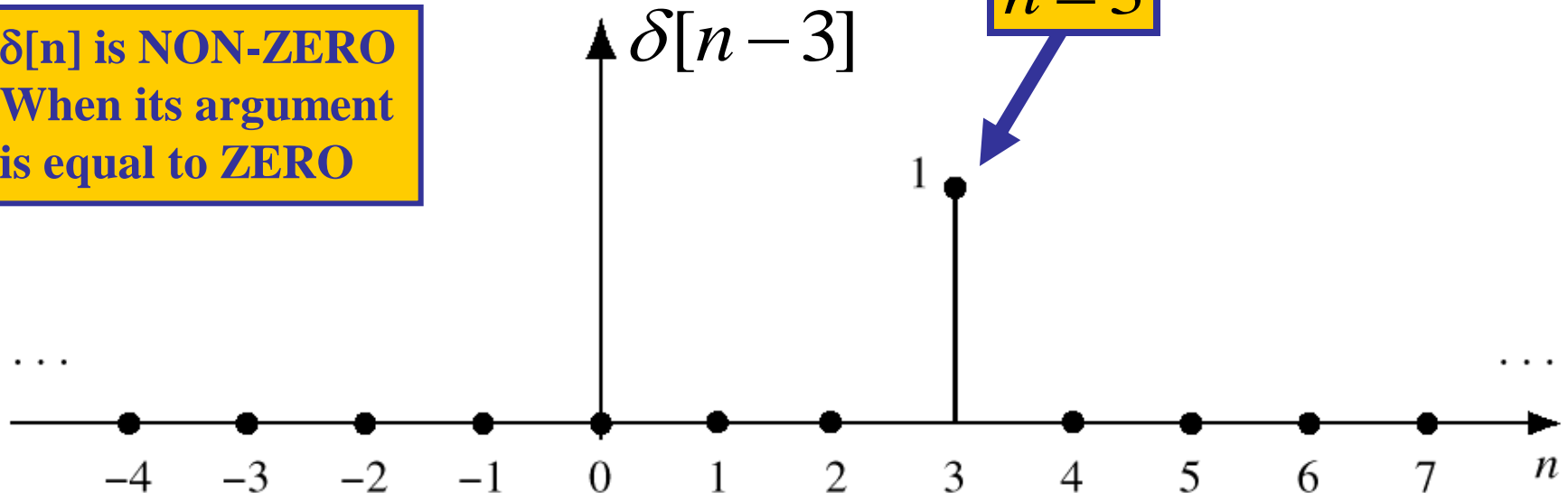
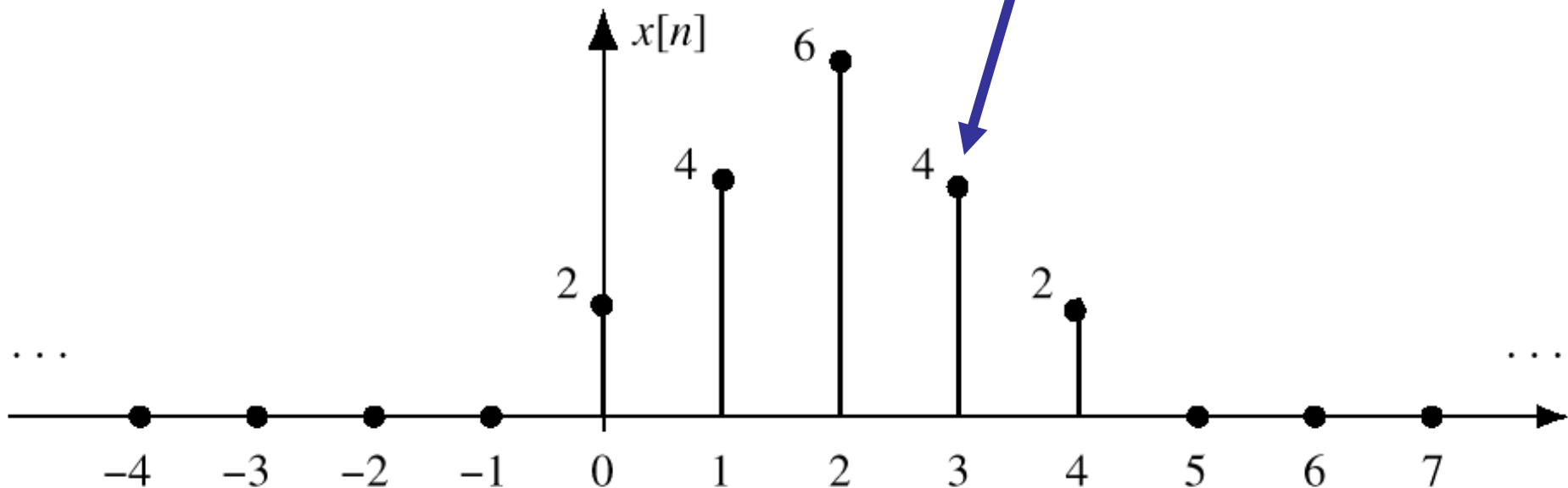


Figure 5.7 Shifted impulse sequence,  $\delta[n - 3]$ .

# MATH FORMULA for $x[n]$

- Use **SHIFTED** IMPULSES to write  $x[n]$

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



# SUM of 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

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

This formula ALWAYS works

$$= \dots + x[-1]\delta[n + 1] + x[0]\delta[n] + x[1]\delta[n - 1] + \dots \quad (5.3.6)$$

# 4-pt AVERAGER

- CAUSAL SYSTEM: USE PAST VALUES

$$y[n] = \frac{1}{4} (x[n] + x[n-1] + x[n-2] + x[n-3])$$

- INPUT = UNIT IMPULSE SIGNAL =  $\delta[n]$

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

$$y[n] = \frac{1}{4} \delta[n] + \frac{1}{4} \delta[n-1] + \frac{1}{4} \delta[n-2] + \frac{1}{4} \delta[n-3]$$

- OUTPUT is called “IMPULSE RESPONSE”

$$h[n] = \{ \dots, 0, 0, \frac{1}{4}, \frac{1}{4}, \frac{1}{4}, \frac{1}{4}, 0, 0, \dots \}$$

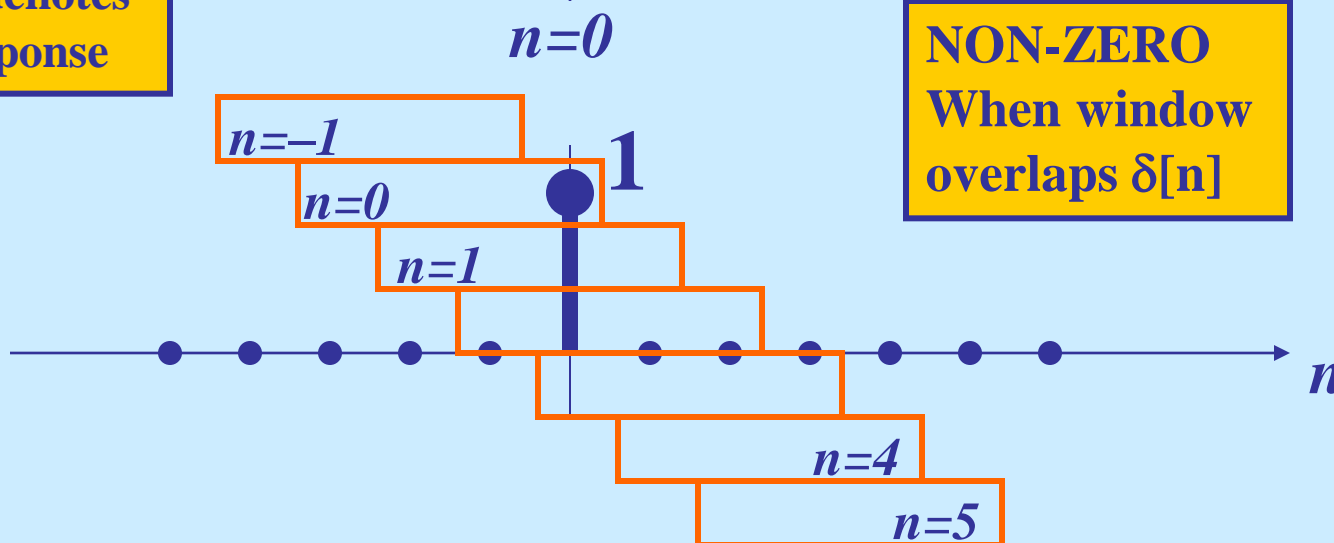
# 4-pt Avg Impulse Response

$$y[n] = \frac{1}{4} (x[n] + x[n-1] + x[n-2] + x[n-3])$$

$\delta[n]$  “READS OUT” the FILTER COEFFICIENTS

$$h[n] = \{ \dots, 0, 0, \frac{1}{4}, \frac{1}{4}, \frac{1}{4}, \frac{1}{4}, 0, 0, \dots \}$$

“h” in  $h[n]$  denotes  
Impulse Response





# FIR IMPULSE RESPONSE

- Convolution = Filter Definition
  - Filter Coeffs = Impulse Response

$n$	$n < 0$	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

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

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

**CONVOLUTION**

# FILTERING EXAMPLE

- 7-point AVERAGER

- Smooth compared to 3-point Averager
  - By making its amplitude (A) smaller

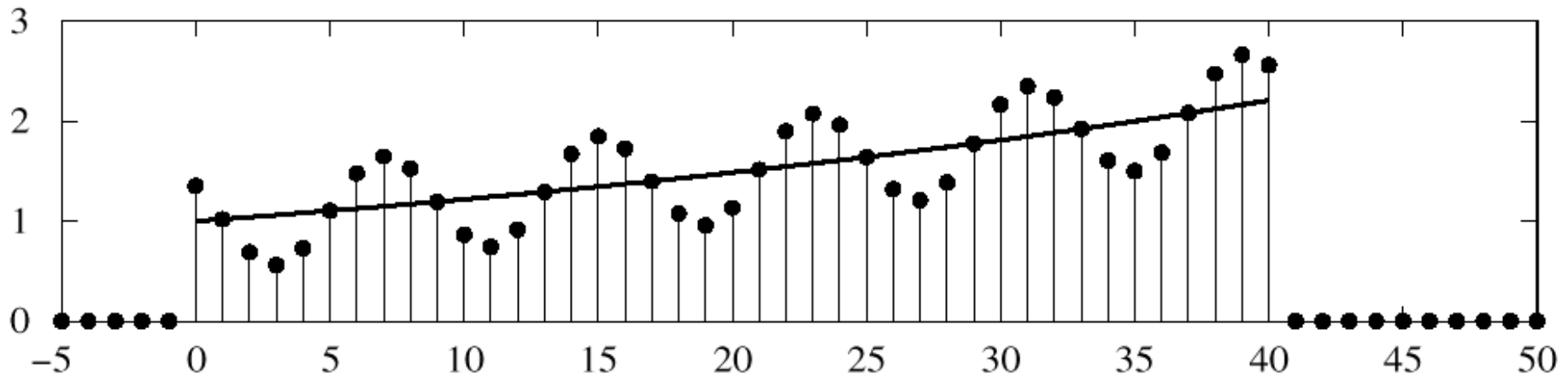
- 3-point AVERAGER
  - Changes A slightly

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

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

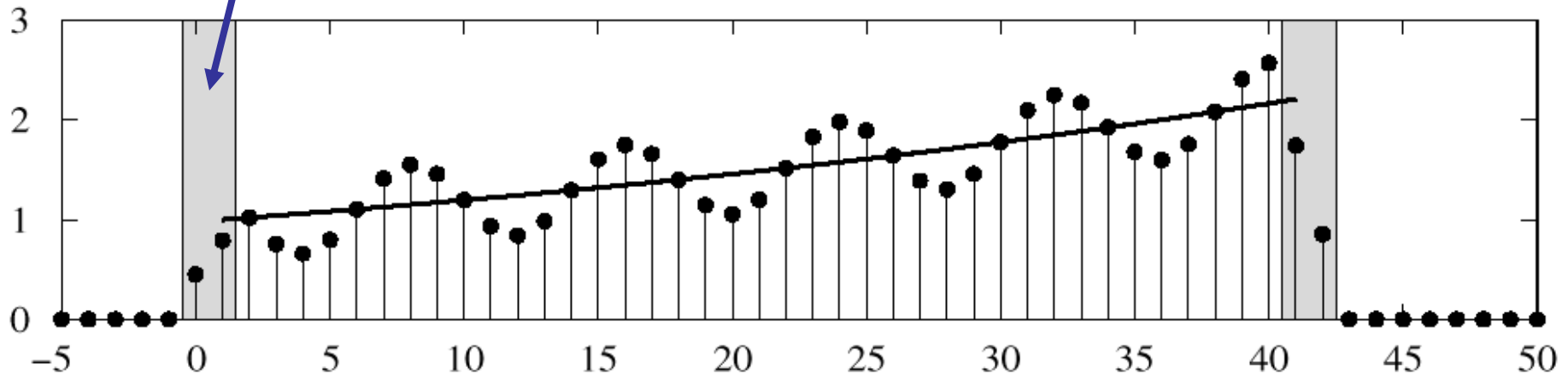
# 3-pt AVG EXAMPLE

Input:  $x[n] = (1.02)^n + \cos(2\pi n/8 + \pi/4)$  for  $0 \leq n \leq 40$



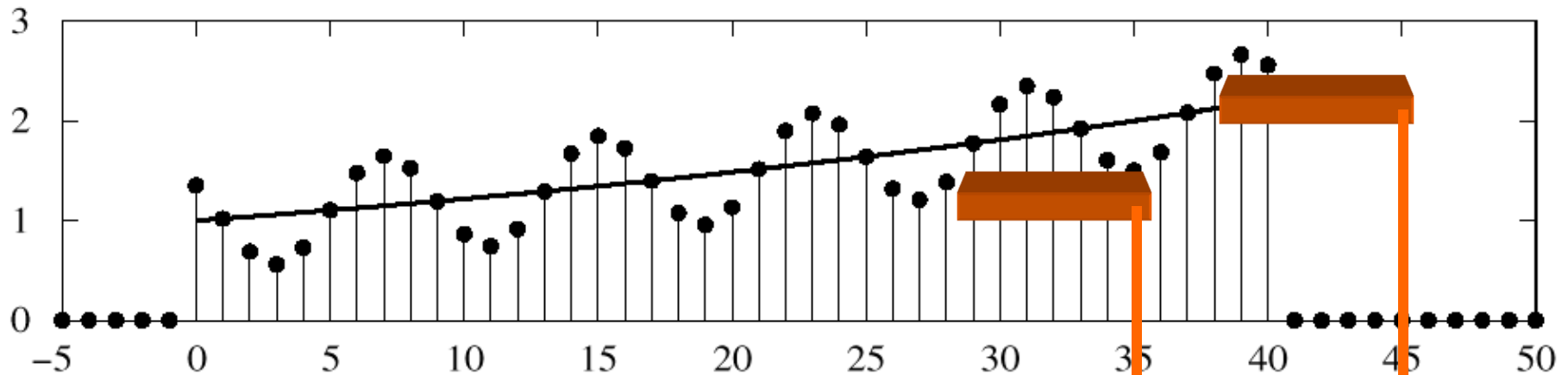
USE PAST VALUES

Output of 3-Point Running-Average Filter



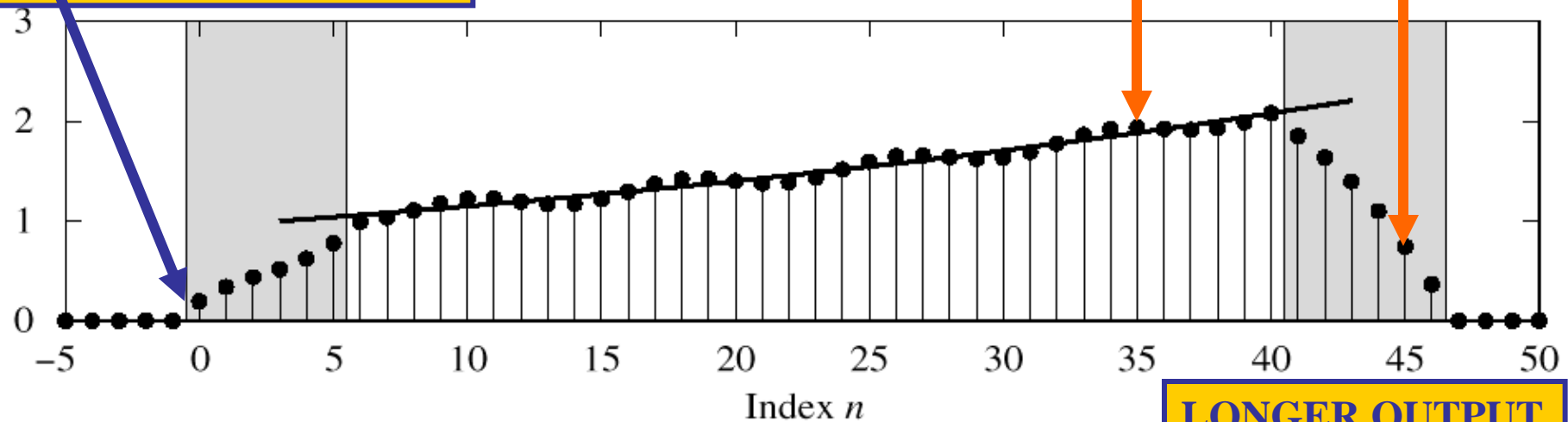
# 7-pt FIR EXAMPLE (AVG)

Input:  $x[n] = (1.02)^n + \cos(2\pi n/8 + \pi/4)$  for  $0 \leq n \leq 40$



**CAUSAL: Use Previous**

Output of 7-Point Running-Average Filter



**LONGER OUTPUT**