

DSP First, 2/e



MODIFIED TLH

Lecture 12

Linearity & Time-Invariance

Convolution

READING ASSIGNMENTS



- This Lecture:
 - Chapter 5, Sections 5-4, 5-6, 5-7 & 5-8
 - Section 5-5 is covered, but not “in depth”
 - Convolution in Section 5-7 is important

LECTURE OBJECTIVES

- GENERAL PROPERTIES of FILTERS
 - LINEARITY
 - TIME-INVARIANCE
 - \implies CONVOLUTION
- BLOCK DIAGRAM REPRESENTATION
 - Components for Hardware
 - Connect Simple Filters Together to Build More Complicated Systems

LTI SYSTEMS

OVERVIEW

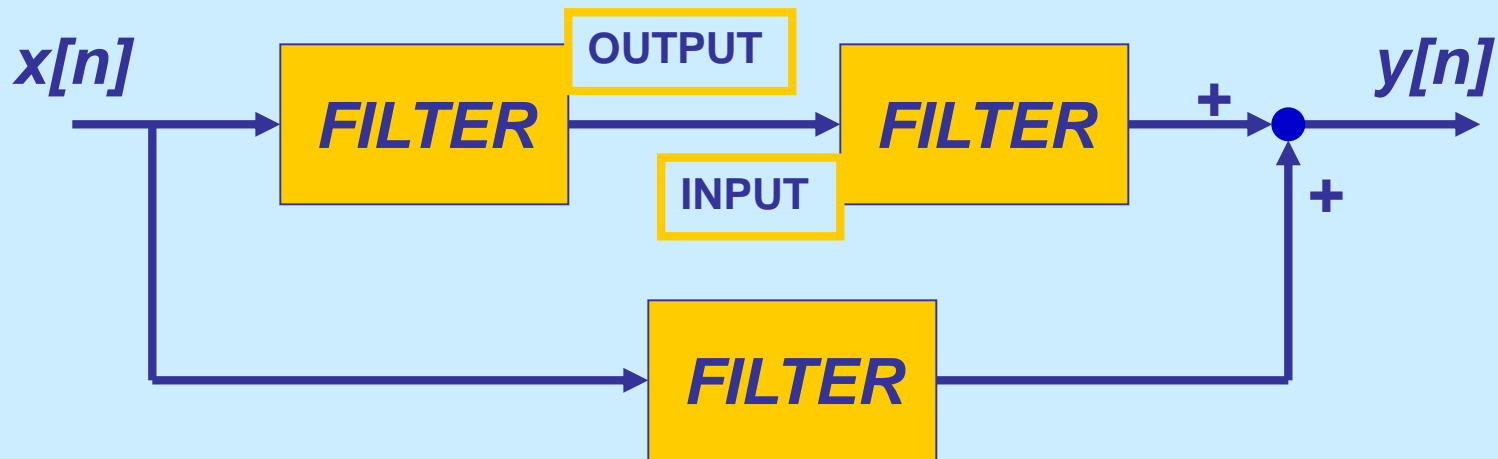
- IMPULSE RESPONSE, $h[n]$
 - FIR case: same as $\{b_k\}$
- CONVOLUTION
 - GENERAL: $y[n] = h[n] * x[n]$
 - GENERAL CLASS of SYSTEMS
 - LINEAR and TIME-INVARIANT
- ALL LTI systems have $h[n]$ & use convolution

DIGITAL FILTERING



- CONCENTRATE on the FILTER (DSP)
- DISCRETE-TIME SIGNALS
 - FUNCTIONS of n , the “time index”
 - INPUT $x[n]$
 - OUTPUT $y[n]$

BUILDING BLOCKS

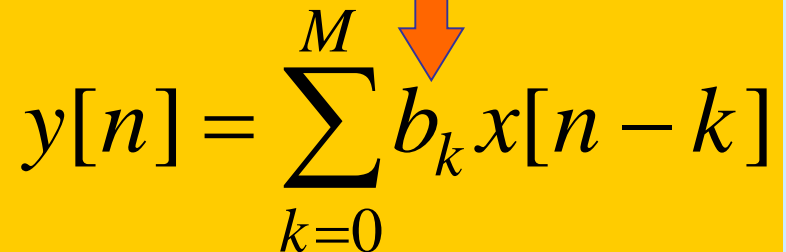


- BUILD UP COMPLICATED FILTERS
 - FROM SIMPLE **MODULES**
 - Ex: FILTER **MODULE** MIGHT BE 3-pt FIR

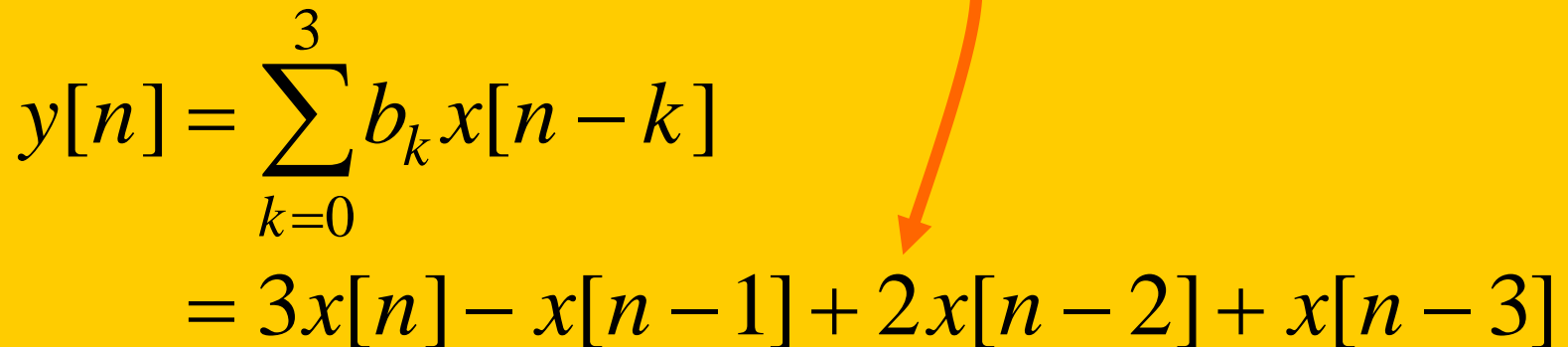
GENERAL FIR FILTER

- FILTER COEFFICIENTS $\{b_k\}$

- DEFINE THE FILTER


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

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


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

MATLAB for FIR FILTER

- $yy = \text{conv}(bb, xx)$
 - VECTOR **bb** contains Filter Coefficients
 - DSP-First: $yy = \text{firfilt}(bb, xx)$
- FILTER COEFFICIENTS $\{b_k\}$

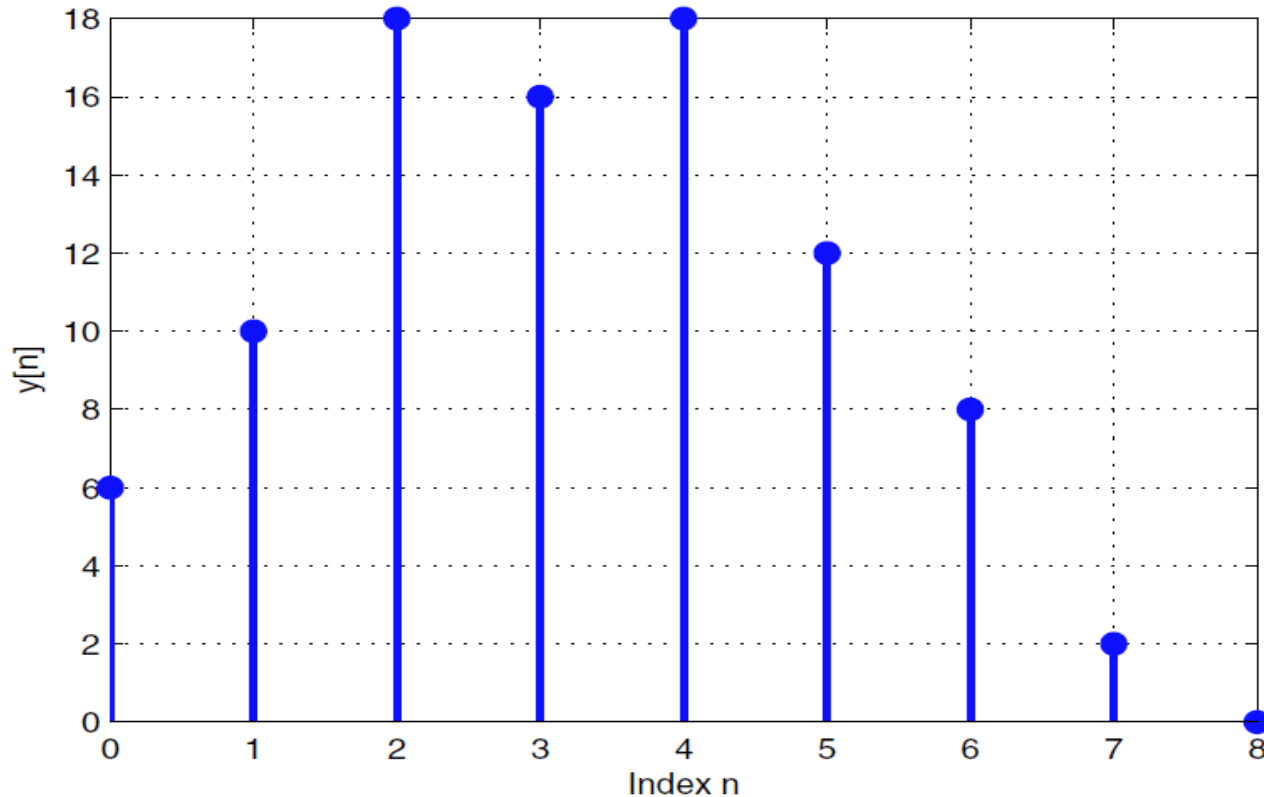
$$y[n] = \sum_{k=0}^M b_k x[n-k]$$

conv2 ()
for images

- We can check the answers using MATLAB's filter function

```
>> n = 0:8;  
>> x = [2 4 6 4 2 0 0 0 0];  
>> h = [3 -1 2 1];  
>> y = filter(h,1,x);  
>> y
```

y = 6 10 18 16 18 12 8 2 0



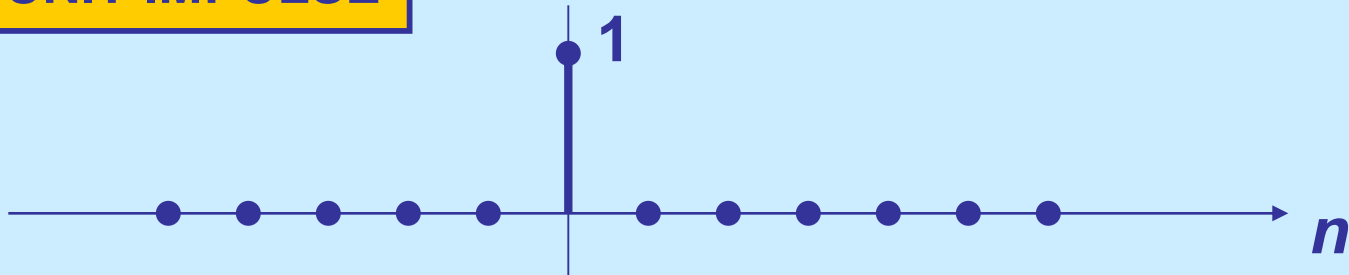
SPECIAL INPUT SIGNALS

- $x[n] = \text{SINUSOID}$
- $x[n]$ has only one **NON-ZERO VALUE**

Later, sinusoid leads to the
FREQUENCY RESPONSE

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

UNIT-IMPULSE



UNIT IMPULSE RESPONSE

- FIR filter DIFFERENCE EQUATION is specified by the filter coefficients b_k

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

- EQUIVALENCE: can we describe the filter using a **SIGNAL** instead?

FIR IMPULSE RESPONSE

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

$$h[n] = \sum_{k=0}^M b_k \delta[n - k]$$

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

- Impulse response $h[k]=b_k$ is, in fact, a **SIGNAL** description of filter coefficients
- Allows us to write **CONVOLUTION** sum

LTI: Convolution Sum

- **Output = Convolution of $x[n]$ & $h[n]$**
 - NOTATION: $y[n] = h[n] * x[n]$
 - FIR case:

FINITE LIMITS

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


Same as b_k

FINITE LIMITS

$$y[n] = h[n] * x[n]$$

LTI: Convolution Sum

- Delay the signal $x[n]$ & then multiply by filter coefficients that come from $h[n]$


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

$$= h[0]x[n] + h[1]x[n-1] + h[2]x[n-2] + \dots$$

CONVOLUTION Example

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

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

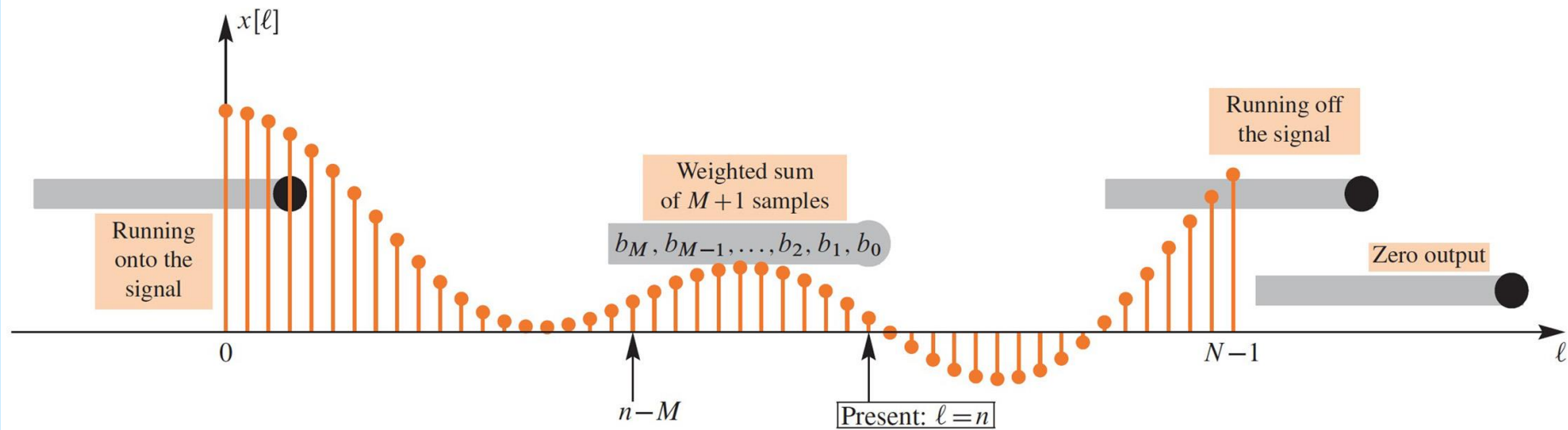
$$x[n] = u[n]$$

n	-1	0	1	2	3	4	5	6	7
$x[n]$	0	1	1	1	1	1	1	1	...
$h[n]$	0	1	-1	2	-1	1	0	0	0
$h[0]x[n]$	0	1	1	1	1	1	1	1	1
$h[1]x[n-1]$	0	0	-1	-1	-1	-1	-1	-1	-1
$h[2]x[n-2]$	0	0	0	2	2	2	2	2	2
$h[3]x[n-3]$	0	0	0	0	-1	-1	-1	-1	-1
$h[4]x[n-4]$	0	0	0	0	0	1	1	1	1
$y[n]$	0	1	0	2	1	2	2	2	...

GENERAL CAUSAL FIR FILTER

- SLIDE a Length-L WINDOW across $x[n]$

$$y[n] = \sum_{k=0}^M b_k x[n-k] = b_0 x[n] + b_1 x[n-1] + \dots + b_M x[n-M]$$



$x[n-M]$

$x[n]$

POP QUIZ

- FIR Filter is “FIRST DIFFERENCE”
 - $y[n] = x[n] - x[n - 1]$
- Write output as a convolution
 - Need impulse response

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

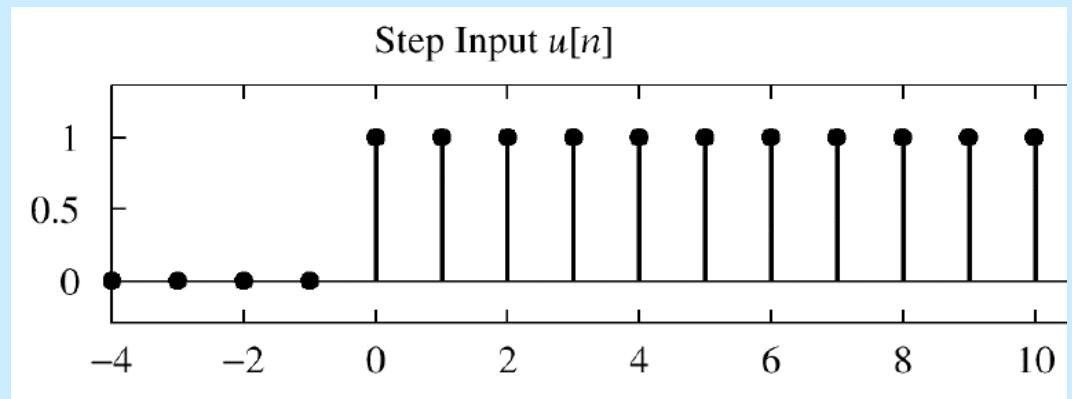
- Then, an equivalent way to compute the output:

$$\begin{aligned} y[n] &= h[n] * x[n] = (\delta[n] - \delta[n - 1]) * x[n] \\ &= (\delta[n] * x[n]) - (\delta[n - 1] * x[n]) \\ &= x[n] - x[n - 1] \end{aligned}$$

POP QUIZ

- FIR Filter is “FIRST DIFFERENCE”
 - $y[n] = x[n] - x[n-1]$
- INPUT is “UNIT STEP”

$$u[n] = \begin{cases} 1 & n \geq 0 \\ 0 & n < 0 \end{cases}$$



- Find $y[n]$?

$$y[n] = u[n] - u[n-1] = \delta[n]$$

HARDWARE STRUCTURES



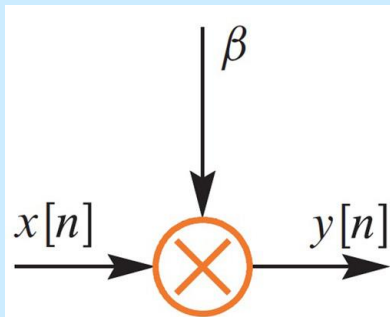
$$y[n] = \sum_{k=0}^M b_k x[n - k]$$

- INTERNAL STRUCTURE of “FILTER”
 - WHAT COMPONENTS ARE NEEDED?
 - HOW DO WE “HOOK” THEM TOGETHER?
- SIGNAL FLOW GRAPH NOTATION

HARDWARE ATOMS

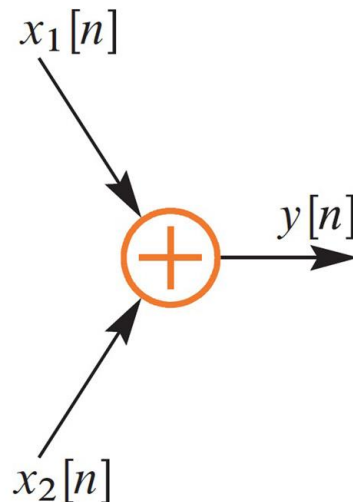
- Add, Multiply & Store

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



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

(a)



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



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

(c)

FIR STRUCTURE

- Direct Form

SIGNAL
FLOW GRAPH

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

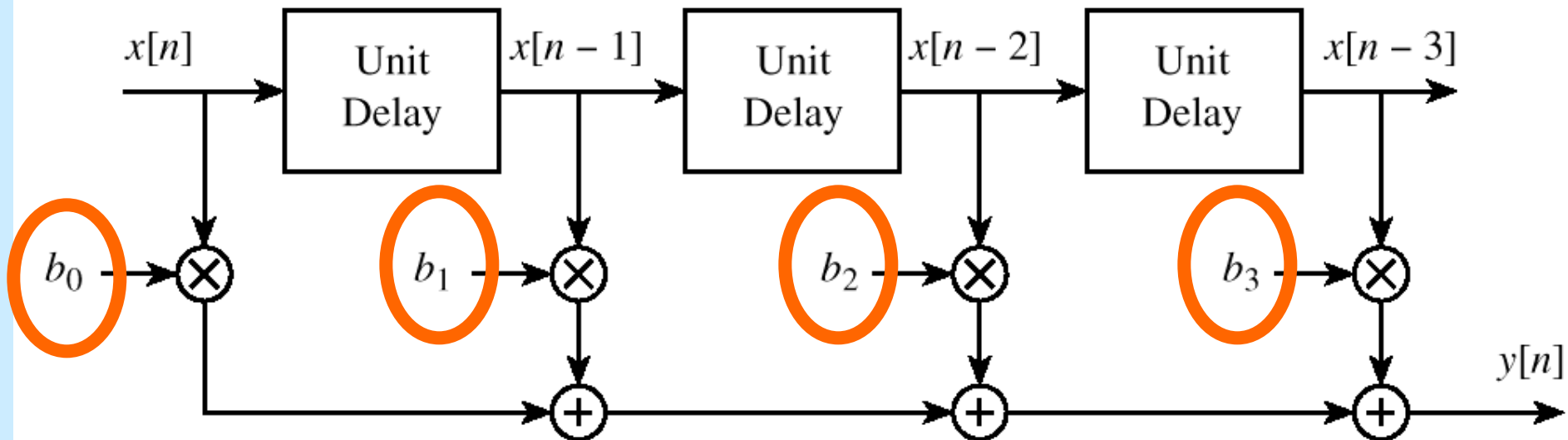


Figure 5.13 Block-diagram structure for the M th order FIR filter.

SYSTEM PROPERTIES



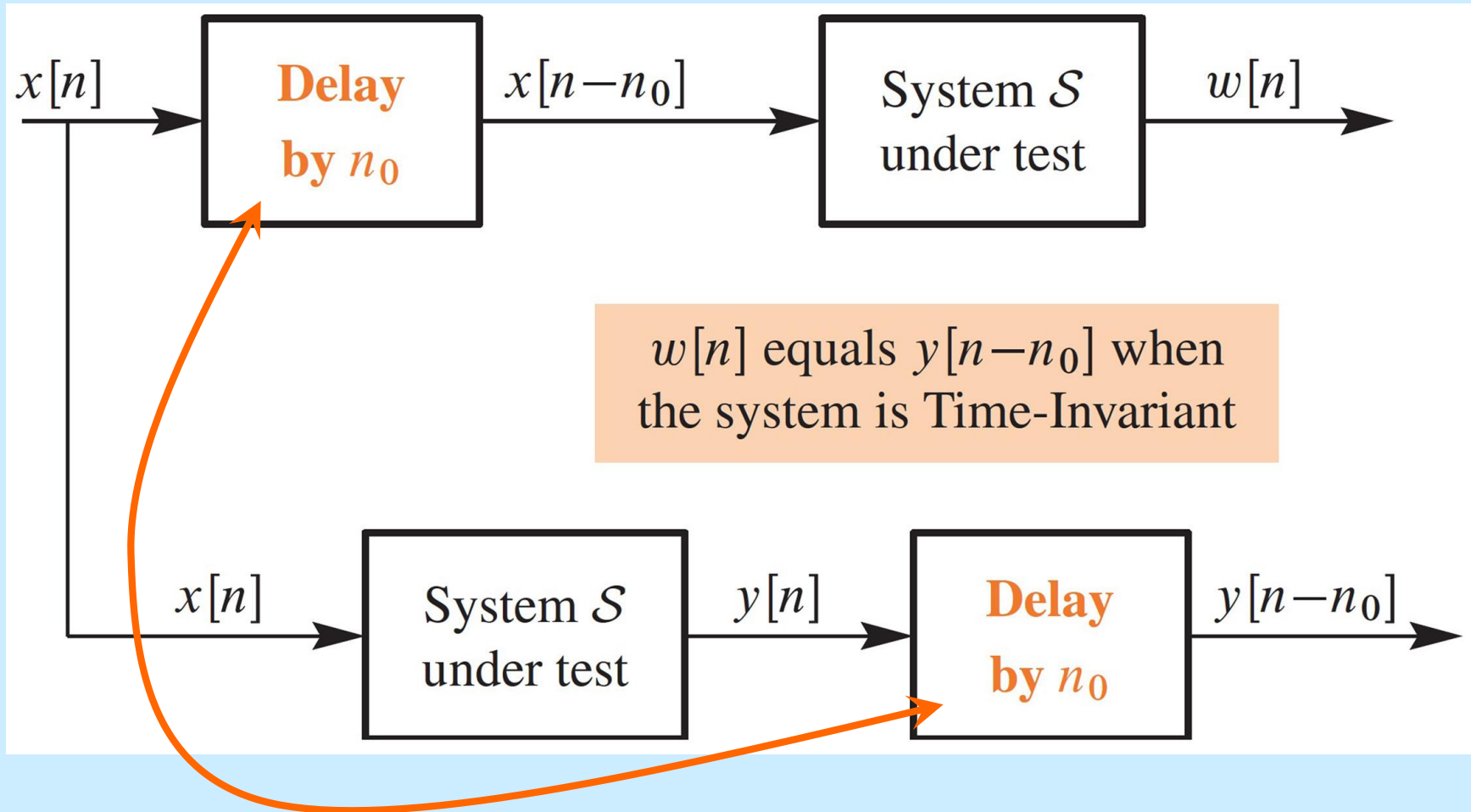
- MATHEMATICAL DESCRIPTION
- TIME-INVARIANCE
- LINEARITY
- CAUSALITY
 - “No output prior to input”

TIME-INVARIANCE



- IDEA:
 - “Time-Shifting the input will cause the **same** time-shift in the output”
- EQUIVALENTLY,
 - We can prove that
 - The time origin ($n=0$) is picked arbitrary

TESTING Time-Invariance

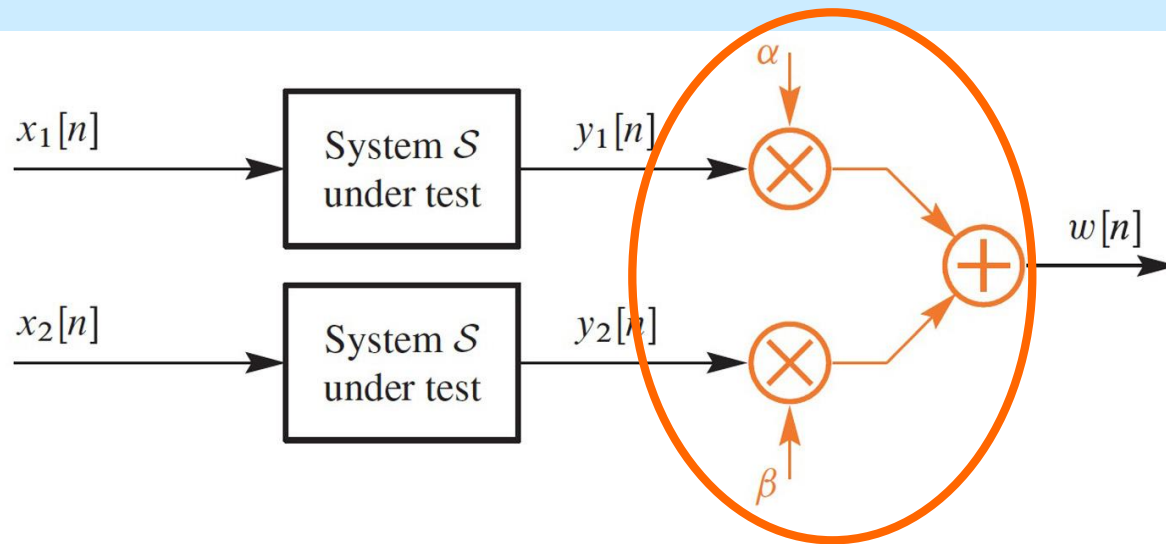


LINEAR SYSTEM

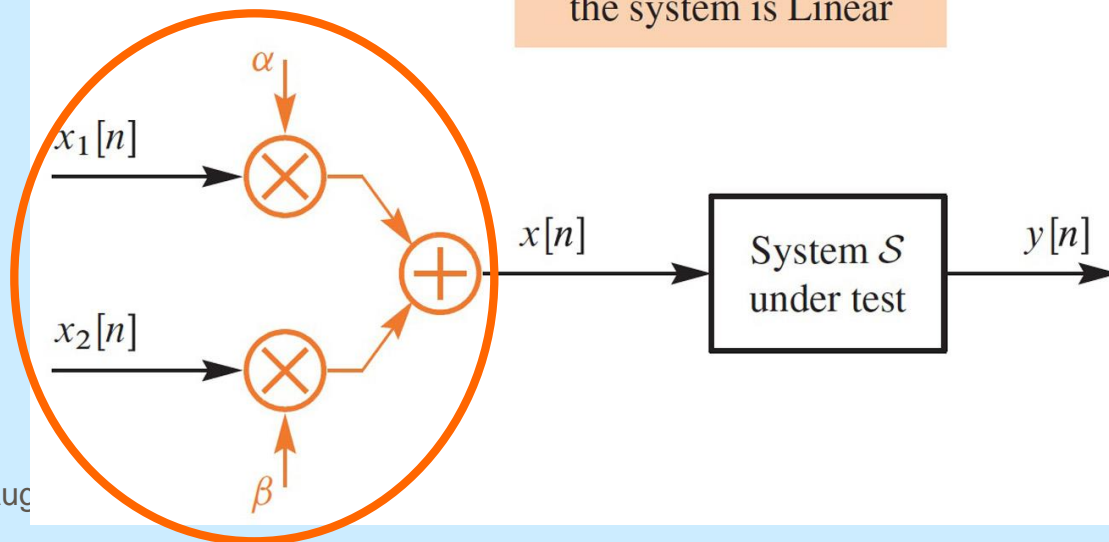


- LINEARITY = Two Properties
- SCALING
 - “Doubling $x[n]$ will double $y[n]$ ”
- SUPERPOSITION:
 - “Adding two inputs gives an output that is the sum of the individual outputs”

TESTING LINEARITY



$w[n]$ equals $y[n]$ when the system is Linear



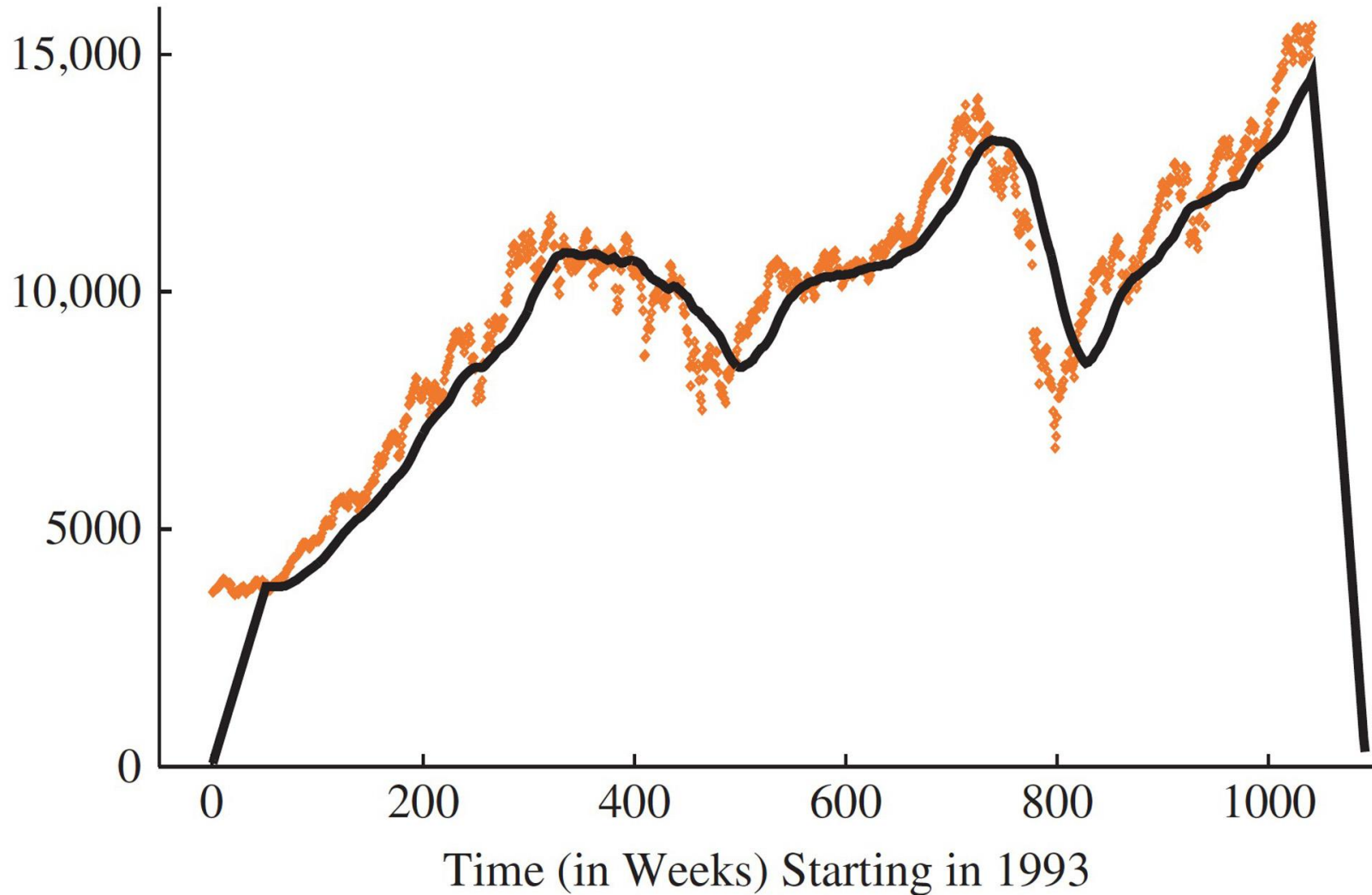
LTI SYSTEMS

$$y[n] = h[n] * x[n]$$

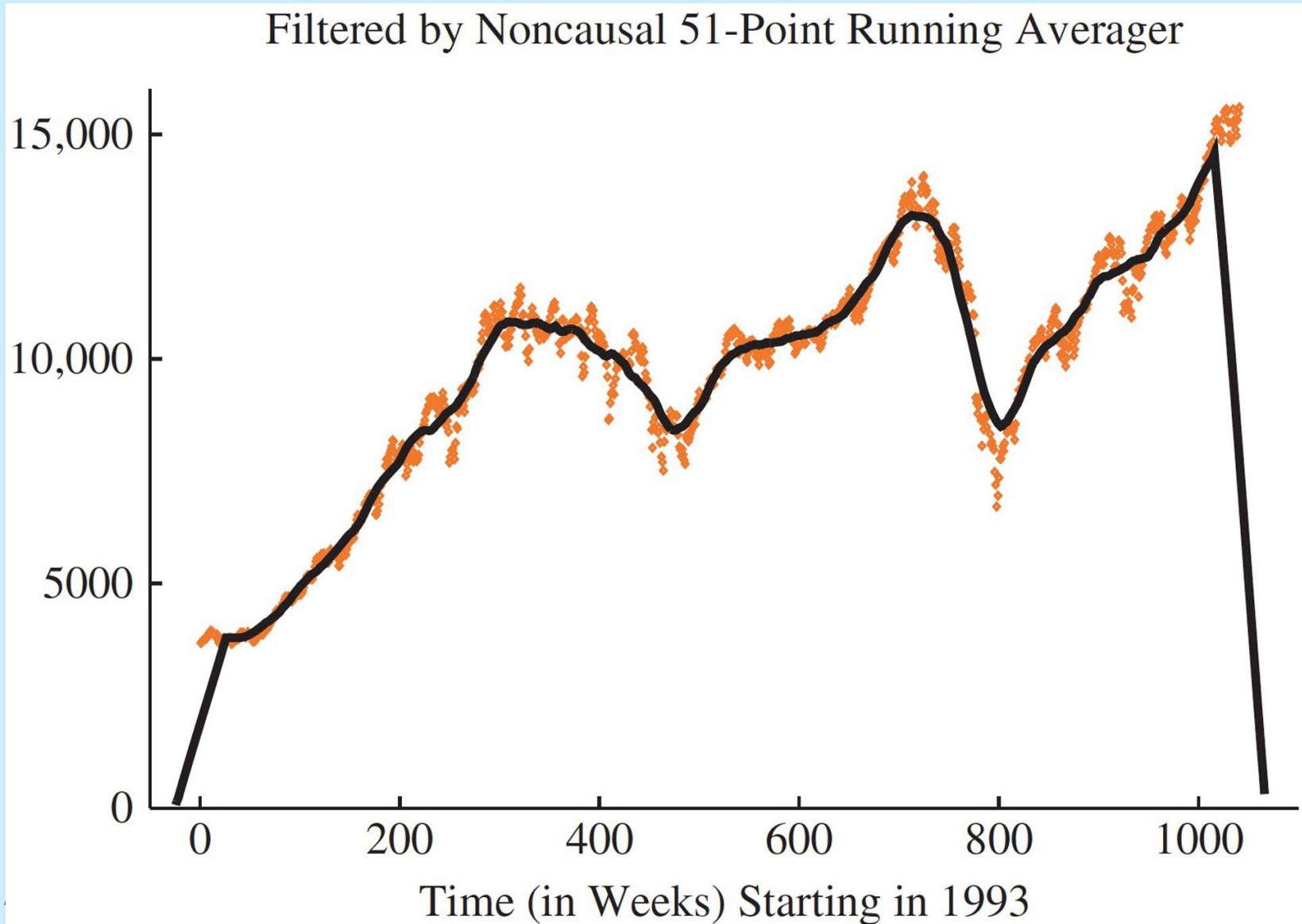
- LTI: **L**inear & **T**ime-**I**nvariant
- COMPLETELY CHARACTERIZED by:
 - IMPULSE RESPONSE $h[n]$
 - CONVOLUTION:
 - The “rule” defining the system can ALWAYS be re-written as convolution
- FIR Example: $h[n]$ is same as b_k

FILTER STOCK PRICES

Filtered by Causal 51-Point Running Averager



STOCK PRICES FILTERED (2)



CASCADE SYSTEMS

- Does the order of S_1 & S_2 matter?
 - NO, LTI SYSTEMS can be rearranged !!!
 - WHAT ARE THE FILTER COEFFS? $\{b_k\}$

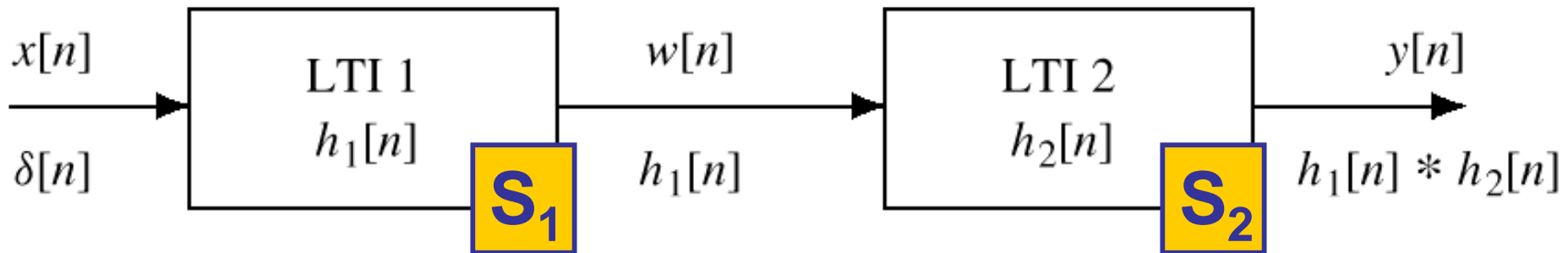


Figure 5.19 A Cascade of Two LTI Systems.