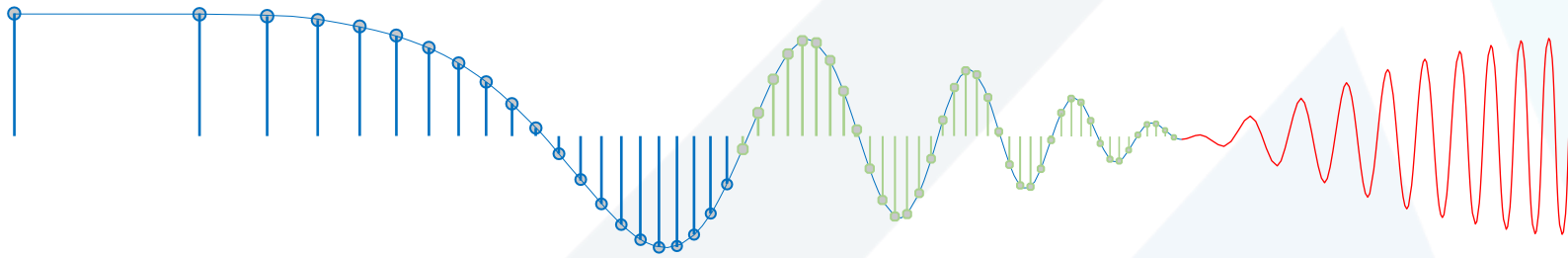


# CEDC606: Digital Signal Processing

## Lecture Notes 2: Discrete-Time Signals and Systems



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## Chapter 2

### Discrete-time signals and systems

1. Discrete-time signals
2. Discrete-time systems
3. Linear time-invariant (LTI) systems
4. Linear constant-coefficient difference equations (LCCDE)

## 1. Discrete-time signals

- A discrete-time signal  $x[n]$  is a **sequence** of numbers defined for every value of the integer variable  $n$ .
- A discrete-time signal is **not defined** for noninteger values of  $n$ . For example, the value of  $x[3/2]$  is not zero, just undefined.
- When  $x[n]$  is obtained by sampling a continuous-time signal  $x(t)$ , the interval  $T_s$  between two successive samples is known as the **sampling period**.
- The quantity  $F_s = 1/T_s$ , called the **sampling frequency**, equals the number of samples per unit of time.
- The **duration** or **length**  $L_x$  of a discrete-time signal  $x[n]$  is the number of samples from the first **nonzero** sample  $x[n_1]$  to the last **nonzero** sample  $x[n_2]$ , that is  $L_x = n_2 - n_1 + 1$ .

- The **range**  $n_1 \leq n \leq n_2$ , denoted by  $[n_1, n_2]$  is called the **support** of the sequence.
- There are several ways to **represent** a DT signal. The more widely used are:

- Functional representation  $x[n] = \begin{cases} (\frac{1}{2})^n, & n \geq 0 \\ 0, & n < 0 \end{cases}$

- Tabular representation
- |        |     |    |    |   |     |     |     |     |
|--------|-----|----|----|---|-----|-----|-----|-----|
| $n$    | ... | -2 | -1 | 0 | 1   | 2   | 3   | ... |
| $x[n]$ | ... | 0  | 0  | 1 | 1/2 | 1/4 | 1/8 | ... |

- Sequence representation  $x[n] = \{\dots, 0, \underset{\uparrow}{1}, 1/2, 1/4, 1/8, \dots\}$

The **symbol**  $\uparrow$  denotes the index  $n = 0$ ; it is omitted when the table starts at  $n = 0$ .

- Graphical representation
- 

- The energy of a sequence  $x[n]$  is defined by: 
$$E_x = \sum_{n=-\infty}^{\infty} |x[n]|^2$$
- The power of a sequence  $x[n]$  is defined by: 
$$P_x = \lim_{L \rightarrow \infty} \left[ \frac{1}{2L + 1} \sum_{n=-L}^L |x[n]|^2 \right]$$

## Elementary discrete-time signals

- Unit impulse sequence: 
$$\delta[n] = \begin{cases} 1 & n = 0 \\ 0, & n \neq 0 \end{cases}$$
- Unit step sequence: 
$$u[n] = \begin{cases} 1 & n \geq 0 \\ 0, & n < 0 \end{cases}$$
- Real sinusoidal sequence:  $x[n] = A \cos(\omega_0 n + \phi), \quad -\infty < n < \infty$   
 where  $A$  (amplitude),  $\omega_0$  (frequency) and  $\phi$  (phase) are real constants.

- Exponential sequence:  $x[n] = A a^n$ ,  $-\infty < n < \infty$   
where  $A$  and  $a$  can take real or complex values.
- If both  $A$  and  $a$  are **real** then  $x[n]$  is termed as a **real exponential sequence**.
  - If  $|a| > 1$ , the magnitude of  $x[n]$  **increases exponentially** as  $n$  increases.
  - If  $|a| < 1$ , the magnitude of  $x[n]$  **decreases exponentially** as  $n$  increases.
  - If  $|a| = 1$ , the magnitude of  $x[n]$  is a **constant**, independent of  $n$ .
  - The values of  $x[n]$  **alternate** in sign when  $a$  is negative.
- If  $A = |A| e^{j\phi}$  and  $a = e^{j\omega_0}$ , then  $x[n]$  is termed as a **complex sinusoid sequence**.

$$x[n] = \underbrace{|A| \cos(\omega_0 n + \phi)}_{\text{Re}\{x[n]\}} + j \underbrace{|A| \sin(\omega_0 n + \phi)}_{\text{Im}\{x[n]\}}$$

Thus  $\text{Re}\{x[n]\}$  and  $\text{Im}\{x[n]\}$  are **real sinusoids**.

- If both  $A = |A|e^{j\phi}$  and  $a = |a|e^{j\omega_0}$  are complex numbers, then:

$$x[n] = \underbrace{|A||a|^n \cos(\omega_0 n + \phi)}_{\text{Re}\{x[n]\}} + j \underbrace{|a|^n |A| \sin(\omega_0 n + \phi)}_{\text{Im}\{x[n]\}}$$

Thus  $\text{Re}\{x[n]\}$  and  $\text{Im}\{x[n]\}$  are each the product of a **real exponential** and **real sinusoid**.

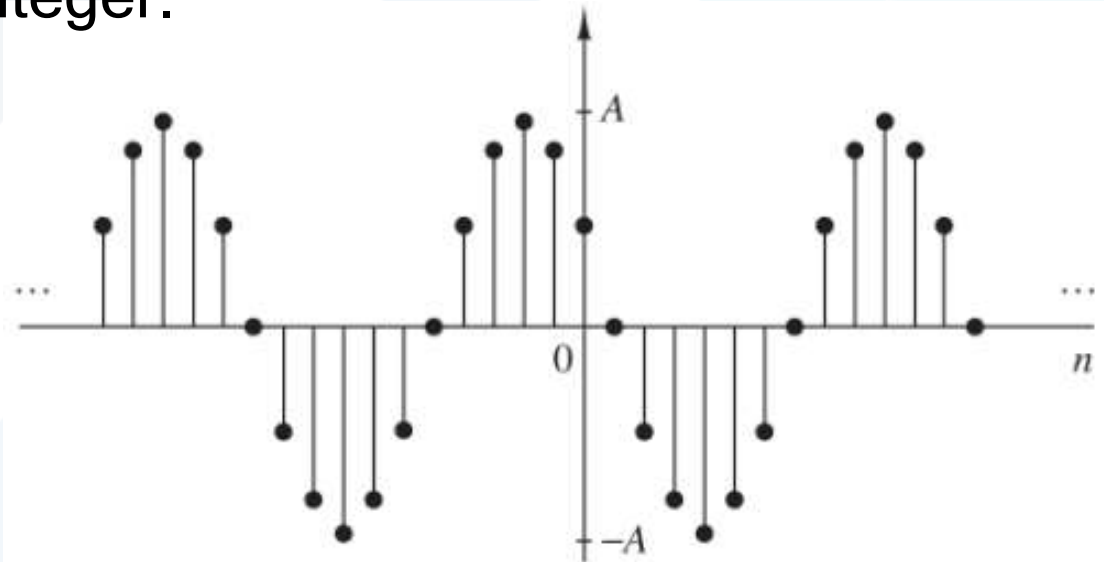
- If  $|a| > 1$   $\text{Re}\{x[n]\}$  and  $\text{Im}\{x[n]\}$  are the product of a **real sinusoid** and a **growing real exponential**.
- If  $|a| < 1$ ,  $\text{Re}\{x[n]\}$  and  $\text{Im}\{x[n]\}$  are the product of a **real sinusoid** and a **decaying real exponential**.
- If  $|a| = 1$ ,  $\text{Re}\{x[n]\}$  and  $\text{Im}\{x[n]\}$  are **real sinusoids**.

- A sequence  $x[n]$  is called **periodic** if  $x[n] = x[n + N]$ , all  $n$ . The smallest value of  $N$  is known as the **fundamental period** or simply period of  $x[n]$ .
- The sinusoidal sequence  $\cos(\omega_0 n + \phi)$  is periodic, if  $\cos(\omega_0 n + \phi) = \cos(\omega_0 n + \omega_0 N + \phi)$ . This is possible if  $\omega_0 N = 2\pi k$ , where  $k$  is an integer ( $\omega_0/2\pi$  is a **rational number**). Therefore the fundamental period is the smallest integer of the form  $2\pi k/\omega_0$ , where  $k$  is a positive integer.

$$x[n] = A \cos\left(\frac{\pi}{6} n + \frac{\pi}{3}\right)$$

$$N = \frac{2\pi k}{\pi/6} = 12k \Rightarrow N = 12 \text{ (for } k = 1\text{)}$$

$$\text{Delay} = 12 \times \frac{\pi/3}{2\pi} = 2 \text{ sampling intervals}$$





## 2. Discrete-time systems

- A discrete-time system is a **computational process** or **algorithm** that transforms or maps a sequence  $x[n]$ , called the input signal, into another sequence  $y[n]$ , called the output signal.
  - $y[n] = \frac{1}{3} \{x[n] + x[n - 1] + x[n - 2]\}$  three-point moving average filter
  - $y[n] = \text{median}\{x[n - 1], x[n - 2], x[n], x[n + 1] + x[n + 2]\}$
- A system is called **causal** if the present value of the output does not depend on future values of the input, that is,  $y[n_0]$  is determined by the values of  $x[n]$  for  $n \leq n_0$ , only.
- A system is said to be **stable**, in the Bounded-Input Bounded-Output (BIBO) sense, if every bounded input signal results in a bounded output signal, that is

$$|x[n]| \leq M_x < \infty \Rightarrow |y[n]| \leq M_y < \infty$$

The three-point moving average filter is stable

$$|x[n]| \leq M_x \Rightarrow |y[n]| \leq |x[n]| + |x[n-1]| + |x[n-2]| = 3M_x = M_y$$

The accumulator system defined by  $y[n] = \sum_{k=0}^{\infty} x[n-k]$

is unstable because the bounded input  $x[n] = u[n]$  produces the output  $y[n] = (n+1)u[n]$ , which becomes unbounded as  $n \rightarrow \infty$ .

- A system  $\mathcal{T}$  is **linear**, if for all functions  $x_1[n]$  and  $x_2[n]$  and all complex constants  $\alpha$  and  $\beta$ , the following condition holds:

$$\mathcal{T}\{\alpha x_1[n] + \beta x_2[n]\} = \alpha \mathcal{T}\{x_1[n]\} + \beta \mathcal{T}\{x_2[n]\}$$

$y[n] = x^2[n]$  is nonlinear system.

- An important consequence of **linearity** is that a linear system cannot produce an output without being **excited**.  $\mathcal{T}\{x[n] = 0\} = y[n] = 0$
- A system  $\mathcal{T}$  is said to be **time invariant** (TI) if, for every function  $x[n]$  and every integer constant  $n_0$ , the following condition holds:

$$\mathcal{T}\{x[n]\} = y[n] \Rightarrow \mathcal{T}\{x[n - n_0]\} = y[n - n_0]$$

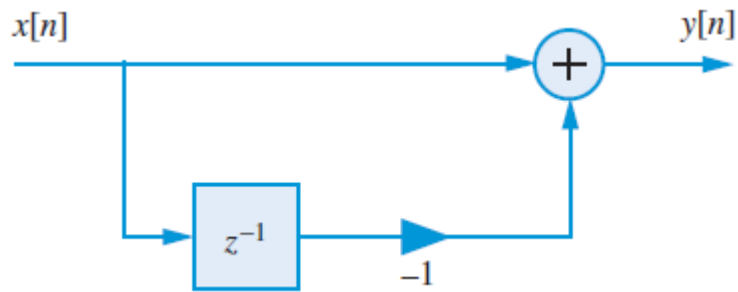
- $y[n] = x[n] \cos \omega_0 n$  is not time invariant system (time-varying).
  - The downsampler system,  $y[n] = \mathcal{T}\{x[n]\} = x[nM]$  is linear but time-varying,
- A system  $\mathcal{T}$  is referred to as **memoryless** if the output  $y[n]$  at every value of  $n$  depends only on the input  $x[n]$  at the same value of  $n$ . Otherwise it is said to be **dynamic**.

$$y[n] = x^2[n] \text{ is a memoryless system.}$$

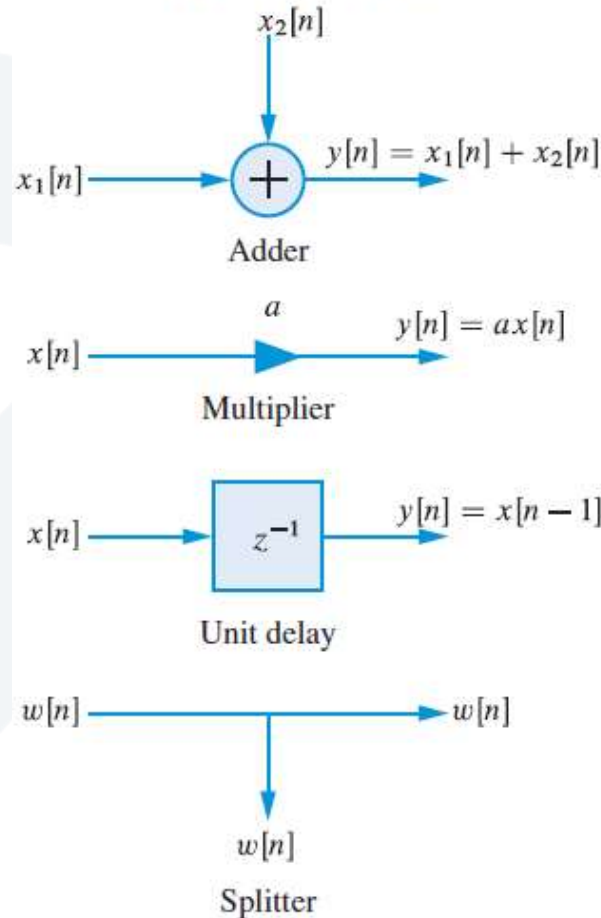
## Block Diagram Representation of Discrete-Time Systems

- **Basic building blocks** The most widely used operations for a block diagram representation of discrete-time systems are provided by the four elementary discrete-time systems (or building blocks) shown below.
- The **adder**, defined by  $y[n] = x_1[n] + x_2[n]$ , computes the sum of two sequences.
- The **constant multiplier**, defined by  $y[n] = ax[n]$ , produces the product of the input sequence by a constant.
- The basic memory element is the **unit delay** system defined by  $y[n] = x[n - 1]$  and denoted by the  $z^{-1}$ . The unit delay is a memory location which can hold (store) the value of a sample for one sampling interval.
- Finally, the **branching element** is used to distribute a signal value to different branches.

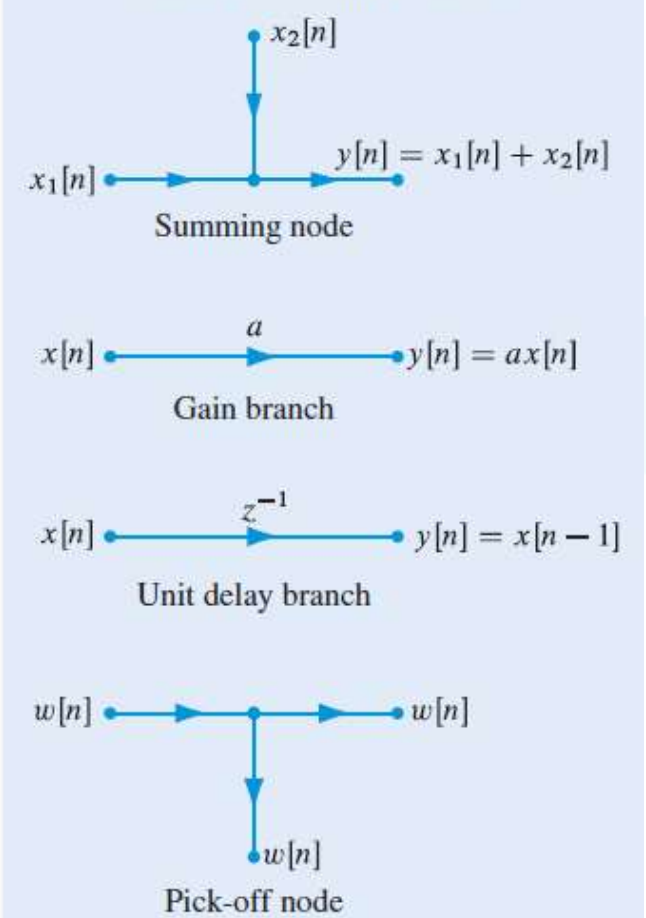
- Figure below shows the block diagram of a system which computes the first difference  $y[n] = x[n] - x[n - 1]$  of its input.



Block Diagram Elements

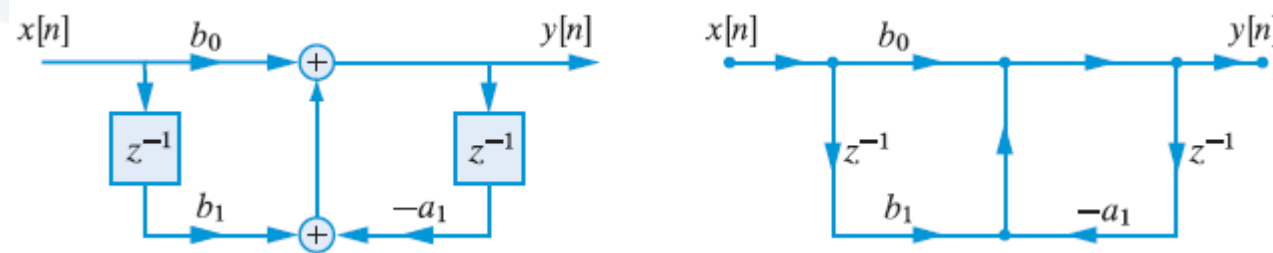


Signal Flow Graph Elements



*Basic building blocks and the corresponding signal flow graph elements*

- To illustrate these concepts, we consider a **first-order** IIR system described by:  $y[n] = b_0x[n] + b_1x[n - 1] - a_1y[n - 1]$ .

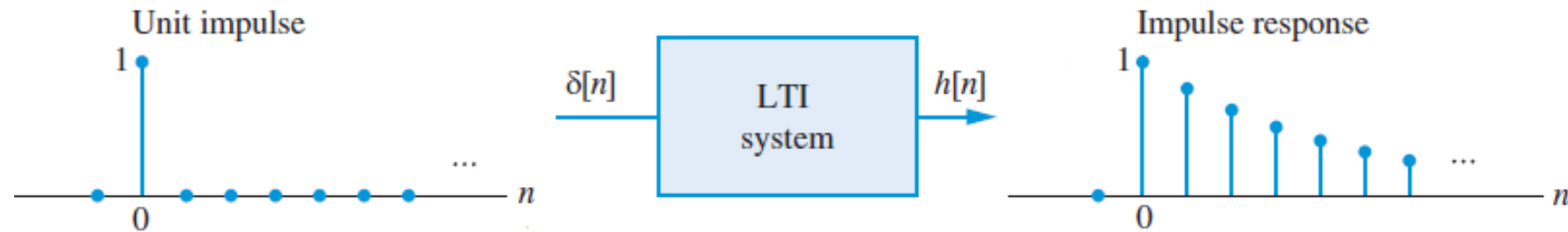


*Structure for the first-order IIR system in block diagram, and signal flow graph*

- A discrete-time system is called **practically realizable** if its practical implementation requires:
  - (1) a **finite** amount of **memory** for the **storage** of signal samples and system parameters, and
  - (2) a **finite** number of **arithmetic** operations for the computation of each output sample.

### 3. Linear time-invariant (LTI) systems

- The **response** of a linear time-invariant (LTI) system to any input can be determined from its response  $h[n]$  to the unit sample sequence  $\delta[n]$ .



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

For example, the unit step can be written as:  $u[n] = \sum_{k=0}^{\infty} \delta[n - k] = \sum_{k=-\infty}^n \delta[k]$

$$y[n] = \mathcal{T} \left\{ \sum_{k=-\infty}^{\infty} x[k] \delta[n - k] \right\} = \sum_{k=-\infty}^{\infty} x[k] \mathcal{T} \{ \delta[n - k] \} = \sum_{k=-\infty}^{\infty} x[k] h_k[n]$$

$h_k[n]$  be the response of the system to the input  $\delta[n - k]$

The property of **time invariance** implies that if  $h[n]$  is the response to  $\delta[n]$ , then the response to  $\delta[n - k]$  is  $h[n - k]$ .

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

This equation is referred to as the **convolution sum**,  $y[n] = x[n] * h[n]$

- **Example 1:** Compute the output  $y[n]$  of a LTI system when:

$$x[n] = \{ \underset{\uparrow}{1}, 2, 3, 4, 5 \}, \quad h[n] = \{ -1, \underset{\uparrow}{2}, 1 \}$$

$$y[-1] = x[0]h[-1] = (1)(-1) = -1$$

$$y[0] = x[0]h[0] + x[1]h[-1] = (1)(2) + (2)(-1) = 0, \quad \dots$$

$$y[n] = \{ -1, \underset{\uparrow}{0}, 2, 4, 6, 14, 5 \}$$



$k$	-3	-2	-1	0	1	2	3	4	5	6	7
$x[k]$				1	2	3	4	5			
$h[k]$			-1	2	1						
$h[-1-k]$		1	2	-1							
$h[-k]$			1	2	-1						
$h[1-k]$				1	2	-1					
$h[2-k]$					1	2	-1				
$h[3-k]$						1	2	-1			
$h[4-k]$							1	2	-1		
$h[5-k]$								1	2	-1	
$y[n]$			-1	0	2	4	6	14	5		
$n$	-3	-2	-1	0	1	2	3	4	5	6	7

$$y[3] = 1 \times 3 + 2 \times 4 - 1 \times 5 = 6$$

*The computation of convolution in tabular form*

- Convolution using direct method

$k$	-3	-2	-1	0	1	2	3	4	5	6	7
$x[k]$				1	2	3	4	5			
$h[k]$			-1	2	1						
				2	4	6	8	10			
					1	2	3	4	5		
			-1	-2	-3	-4	-5				
$y[n]$			-1	0	2	4	6	14	5		
$n$	-3	-2	-1	0	1	2	3	4	5	6	7

x

+

Computation of the convolution sum, the approach is similar to a **pencil** and **paper** multiplication calculation, **except** carries are not performed out of a column.

- Convolution using matrix-vector multiplication

$$\begin{bmatrix} y[-1] \\ y[0] \\ y[1] \\ y[2] \\ y[3] \\ y[4] \\ y[5] \end{bmatrix} = \begin{bmatrix} x[0] & 0 & 0 \\ x[1] & x[0] & 0 \\ x[2] & x[1] & x[0] \\ x[3] & x[2] & x[1] \\ x[4] & x[3] & x[2] \\ 0 & x[4] & x[3] \\ 0 & 0 & x[4] \end{bmatrix} \begin{bmatrix} h[-1] \\ h[0] \\ h[1] \end{bmatrix} = \begin{bmatrix} 1 & 0 & 0 \\ 2 & 1 & 0 \\ 3 & 2 & 1 \\ 4 & 3 & 2 \\ 5 & 4 & 3 \\ 0 & 5 & 4 \\ 0 & 0 & 5 \end{bmatrix} \begin{bmatrix} -1 \\ 2 \\ 1 \end{bmatrix} = \begin{bmatrix} -1 \\ 0 \\ 2 \\ 4 \\ 6 \\ 14 \\ 5 \end{bmatrix}$$

The matrix form of convolution involves a matrix known as **Toeplitz**.

- A simpler approach, from a programming viewpoint, is to express the above equations as a linear combination of column vectors:

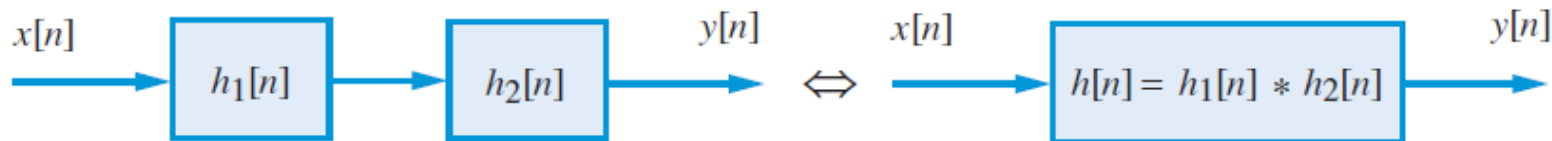
$$\begin{bmatrix} y[-1] \\ y[0] \\ y[1] \\ y[2] \\ y[3] \\ y[4] \\ y[5] \end{bmatrix} = h[-1] \begin{bmatrix} x[0] \\ x[1] \\ x[2] \\ x[3] \\ x[4] \\ 0 \\ 0 \end{bmatrix} + h[0] \begin{bmatrix} 0 \\ x[0] \\ x[1] \\ x[2] \\ x[3] \\ x[4] \\ 0 \end{bmatrix} + h[1] \begin{bmatrix} 0 \\ 0 \\ x[0] \\ x[1] \\ x[2] \\ x[3] \\ x[4] \end{bmatrix}$$

## Properties of linear time-invariant systems

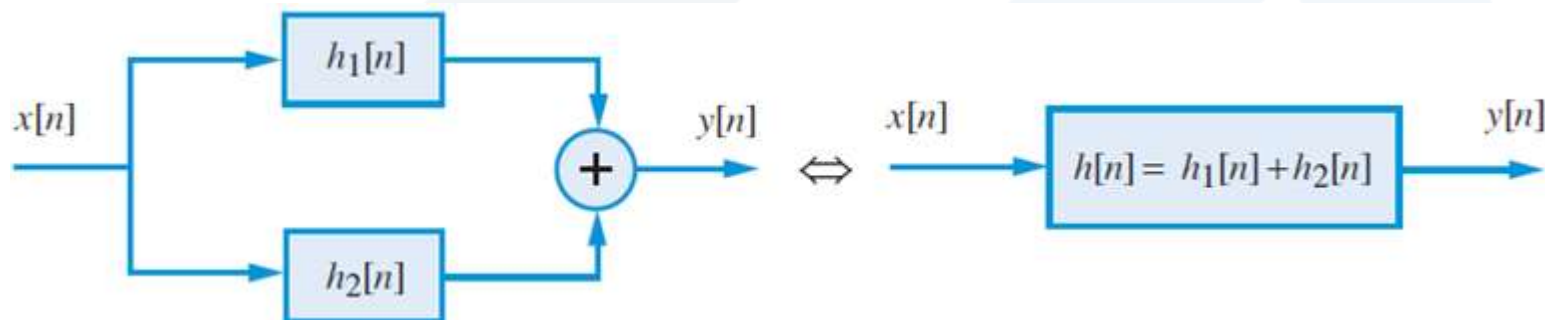
### ▪ Properties of Convolution

- **Convolutional identity:**  $x[n] * \delta[n] = x[n]$
- **Commutative:**  $x[n] * h[n] = h[n] * x[n]$
- **Associative:**  $(x[n] * h_1[n]) * h_2[n] = x[n] * (h_1[n] * h_2[n])$

- **Distributive:**  $x[n] * (h_1[n] + h_2[n]) = x[n] * h_1[n] + x[n] * h_2[n]$
- **Note:** The convolution of two non-periodic sequences:  $x[n]$ ,  $0 \leq n \leq M - 1$  and  $h[n]$ ,  $0 \leq n \leq N - 1$  has length  $M + N - 1$ .
- Cascade interconnection of two LTI systems



- Parallel interconnection of two LTI systems



- Causality and stability

- A LTI system is **causal** if its impulse response  $h[n] = 0$  for  $n < 0$ .
- A LTI system is **stable** if and only if its **impulse response** is absolutely summable,

$$\sum_{n=-\infty}^{\infty} |h[n]| < \infty$$

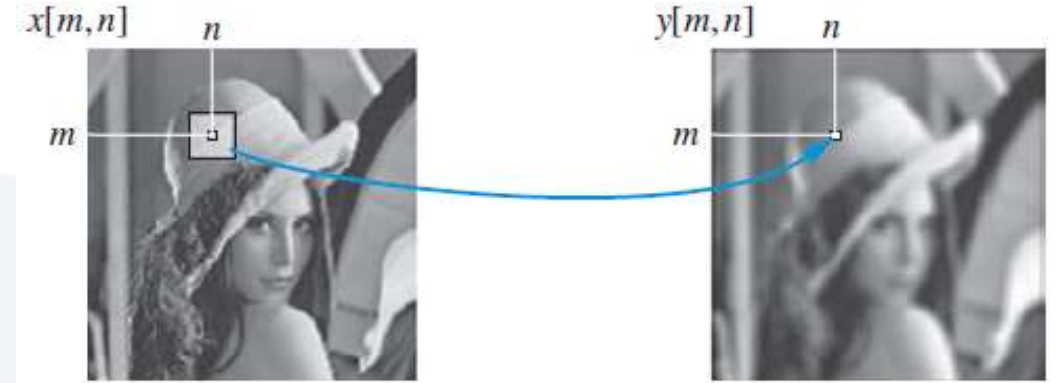
## Convolution in two dimensions

- Spatial filters are very popular and useful in the processing of digital images to implement visual effects like noise filtering, edge detection, etc.
- Smoothing images consists of replacing each pixel by its average over a local region.
- Consider a  $3 \times 3$  region around the pixel  $x[m, n]$ . Then the smoothed pixel value  $y[m, n]$  can be computed as:

$$y[m, n] = \sum_{k=-1}^1 \sum_{l=-1}^1 \left(\frac{1}{9}\right) x[m - k, n - l]$$

We next define a 2D sequence  $h[m, n]$

$$h[m, n] = \begin{cases} \frac{1}{9}, & -1 \leq m, n \leq 1 \\ 0, & \text{otherwise} \end{cases}$$



which can be seen as a spatial filter impulse response:

$$y[m, n] = \sum_{k=-1}^1 \sum_{l=-1}^1 h[k, l] x[m - k, n - l]$$

which is a 2D convolution of image  $x[m, n]$  with a spatial filter  $h[m, n]$ . A general expression for 2D convolution, when the spatial filter has finite symmetric support  $(2K + 1) \times (2L + 1)$ , is given by:

$$y[m, n] = \sum_{k=-K}^K \sum_{l=-L}^L h[k, l] x[m - k, n - l]$$

## 4. Linear constant-coefficient difference equations (LCCDE)

- An important class of LTI systems consists of those systems for which the input  $x[n]$  and the output  $y[n]$  satisfy an  $N$ th-order **linear constant-coefficient difference** equation of the form:

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

- Example 2:** The accumulator system defined by:  $y[n] = \sum_{k=-\infty}^n x[k]$

$$y[n] = x[n] + \sum_{k=-\infty}^{n-1} x[k] = x[n] + y[n-1] \Rightarrow y[n] - y[n-1] = x[n]$$

### Solution of Linear Constant-Coefficient Difference Equations

- The **goal** is to determine the **output**  $y[n]$ ,  $n \geq 0$ , of the system given a **specific input**  $x[n]$ ,  $n \geq 0$ , and a set of **initial conditions**.



- A **solution** to a LCCDE can be obtained in the form:  $y[n] = y_h[n] + y_p[n]$ .

where  $y_h[n]$  is the solution of the **homogeneous linear difference equation** found by setting  $x[n] = 0$ :

$$\sum_{k=0}^N a_k y[n - k] = 0$$

and  $y_p[n]$  is due to the input signal  $x[n]$  being applied to the system. It is referred to as the **particular solution** of the difference equation.

- A **solution** to LCCDE can also be obtained in the form:  $y[n] = y_{zi}[n] + y_{zs}[n]$ .  
where  $y_{zi}[n]$  is called the **zero-input solution**, due to the initial conditions alone (assuming they exist), and  $y_{zs}[n] = h[n] * x[n]$  is called the **zero-state solution**, due to the input  $x[n]$  alone (initial conditions assumed to be zero).
- A **solution** to LCCDE can also be obtained in the form:  $y[n] = y_{tr}[n] + y_{ss}[n]$ .

where  $y_{tr}[n]$  is the **transient response** due to the initial state of the system; It disappears over time, and  $y_{ss}[n]$  is the **steady-state response**; It remains.

- **Example 3:** A causal and stable LTI system

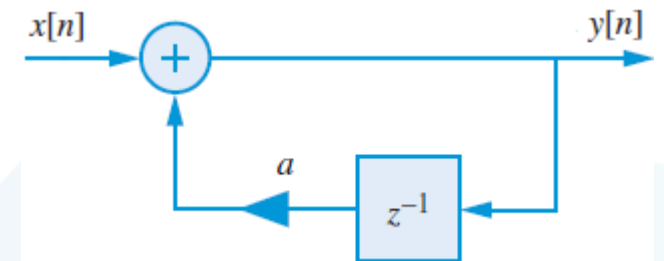
$$y[n] = ay[n - 1] + x[n], \quad |a| < 1$$

We apply an input signal  $x[n]$  to the system for  $n \geq 0$ .

We make no assumptions about the input signal for  $n < 0$ , but we do assume the existence of the initial condition  $y[-1]$ .

Computing successive values of  $y[n]$ :  $y[n] = a^{n+1}y[-1] + \sum_{k=0}^n a^k x[n - k], \quad n \geq 0$

If the system is **initially relaxed** at time  $n = 0$ , then its memory (i.e., the output of the delay) should be zero. Hence  $y[-1] = 0$ . We say that the system is at **zero state** and its corresponding output is called the **zero-state response**,



$$y_{zs}[n] = h[n] * x[n] = \sum_{k=0}^n a^k x[n-k], \quad n \geq 0 \quad \Rightarrow \quad h[n] = a^n u[n]$$

Now, suppose that the system is initially nonrelaxed,  $y[-1] \neq 0$ , and the input  $x[n] = 0$  for all  $n$ . Then the output of the system with zero input is called the

**zero-input response:**  $y_{zi}[n] = y[-1]a^{n+1}, \quad n \geq 0$

$$y[n] = y_{zi}[n] + y_{zs}[n] = \underbrace{y[-1]a^{n+1}}_{\text{zero-input}} + \underbrace{\sum_{k=0}^n a^k x[n-k]}_{\text{zero-state}}, \quad n \geq 0$$

Then, if  $y[-1] = 0$ , the system is LTI. If  $y[-1] \neq 0$ , the system is linear in a more general sense that involves linearity with respect to both input and ICs.

To obtain the step response of the system we set  $x[n] = u[n]$ :

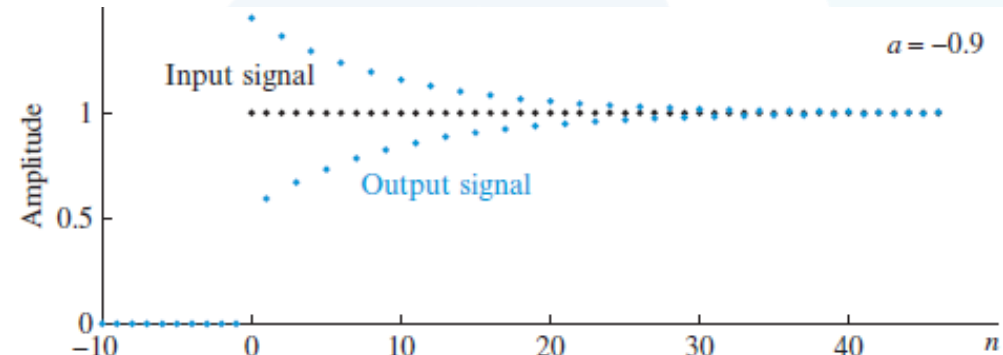
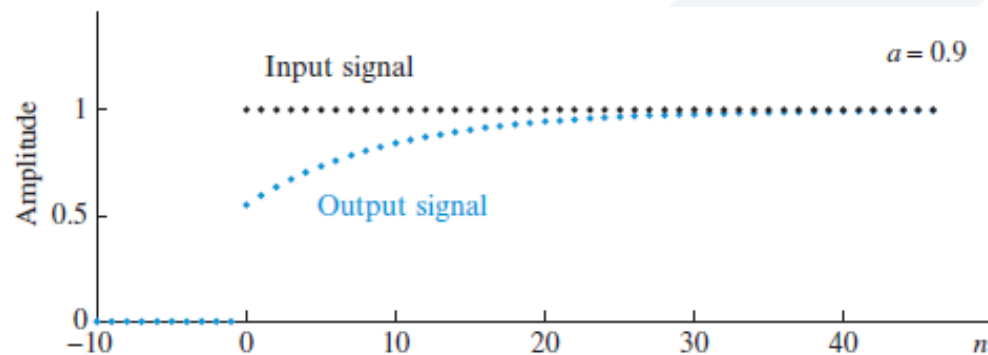
$$y[n] = y_{ss}[n] + y_{tr}[n] = \underbrace{\frac{1}{1-a}}_{\text{steady-state}} + \underbrace{y[-1]a^{n+1} - \frac{a^{n+1}}{1-a}}_{\text{transient}}, \quad n \geq 0$$

For a stable system, that is, when  $|a| < 1$ , we have:

$$y_{ss}[n] = \lim_{n \rightarrow \infty} y[n] = \frac{1}{1-a}, \quad n \geq 0 \qquad y_{tr}[n] = y[-1]a^{n+1} - \frac{a^{n+1}}{1-a} \rightarrow 0$$

$$y[n] = \underbrace{\frac{1}{1-a}}_{\text{steady-state}} + \underbrace{y[-1]a^{n+1} - \frac{a^{n+1}}{1-a}}_{\text{transient}} = \underbrace{\frac{1-a^{n+1}}{1-a}}_{\text{zero-state}} + \underbrace{y[-1]a^{n+1}}_{\text{zero-input}}$$

- **Note:** In general, we have  $y_{zi}[n] \neq y_{tr}[n]$ , and  $y_{ss}[n] \neq y_{zs}[n]$ .
- **Note:** If the system is stable  $y_{ss}[n] = \lim_{n \rightarrow \infty} y_{zs}[n]$ .



## FIR versus IIR systems

- If the unit impulse response of an LTI system is of **finite duration**, then the system is called a **finite-duration impulse response** (or **FIR**) filter.

The following difference equation describes a **causal FIR filter**:

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

- If the impulse response of an LTI system is of infinite duration, then the system is called an **infinite-duration impulse response** (or **IIR**) filter. The following difference equation describes a **recursive IIR filter**:

$$\sum_{k=0}^N a_k y[n - k] = x[n]$$

System	Equation	Linear	Time-invariant	LTI	Causal	Stable
Multiplier	$y[n] = 2x[n]$	✓	✓	✓	✓	✓
Offset	$y[n] = x[n] + 1$	✗	✓	✗	✓	✓
Squarer	$y[n] = x^2[n]$	✗	✓	✗	✓	✓
Delay	$y[n] = x[n - n_0]$	✓	✓	✓	$n_0 \geq 0$	✓
Average	$y[n] = (x[n-1] + x[n] + x[n+1])/3$	✓	✓	✓	✗	✓
Summer	$y[n] = \sum_{k=-\infty}^n x[k]$	✓	✓	✓	✓	✗
LCCDE	$\sum_{k=0}^N a_k y[n - k] = \sum_{k=0}^M b_k x[n - k]$	✓	✓	✓	✓	✓ or ✗
Switch	$y[n] = x[n]u[n]$	✓	✗	✗	✓	✓

*Summary of system properties of example systems*