

# Lecture 4: Time & Frequency Response of Discrete-time System

**ELEN 472: Introduction to Digital Control** 

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## Review

Inverse z-Transform:

$$Y(z) \implies [Inverse z-Transform] \implies y(k)$$

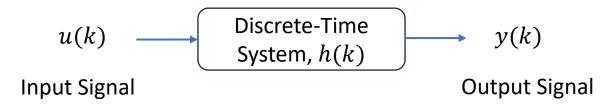
- Long Division
- Partial Fraction Expansion (with a special case with repeated roots)
- Final Value Theorem:
  - Allows us to find the limit of a sequence as time  $k \to \infty$

$$f(\infty) = \lim_{z \to 1} [(z - 1)F(z)]$$

- Z-Transform Solution of Difference Equations:
  - Use z-Transform to solve difference equations
  - Workflow:
    - Use z-Transform to convert difference equations into z-domain;
    - Find expression of the discrete-time signal;
    - Use inverse z-Transform to get the time-domain signal.

## Time Response of Discrete-time Systems

• The **time response** of discrete-time systems is the **output** of the discrete-time system, providing the input.



• To calculate this output signal, we need to use the **convolution**:

$$y(k) = h(k) * u(k) = \sum_{i=0}^{k} h(k-i)u(i)$$

## Example

- Obtain the Convolution of two sequences  $f\{k\}=\{1,1,1\}$  and  $g(k)=\{1,2,3\}$
- Solution:
  - Since  $y(k) = h(k) * u(k) = \sum_{i=0}^{k} h(k-i)u(i)$ .

• 
$$y(k) = f\{k\} * g(k) = \sum_{i=0}^{k} f(k-i)g(i)$$

$$k = 0$$
  $y(0) = f(0).g(0) = 1 \times 1 = 1$   
 $k = 1$   $y(1) = f(1).g(0) + f(0).g(1) = 1 \times 1 + 1 \times 2 = 3$   
 $k = 2$   $y(2) = f(2).g(0) + f(1).g(1) + f(0).g(2) = 1 \times 1 + 1 \times 2 + 1 \times 3 = 6$   
...  $y(3) = f(2).g(1) + f(1).g(2) = 1 \times 2 + 1 \times 3 = 5$   
 $y(4) = f(2).g(2) = 1 \times 3 = 3$   
 $y(k) = 0, k > 4$ 

## **Practice**

- Find the Convolution of two sequences  $f\{k\} = \{2,3,4,5\}$  and  $g(k) = \{1,2,3\}$
- Solution:

• 
$$y(k) = f\{k\} * g(k) = \sum_{i=0}^{k} f(k-i)g(i)$$

• 
$$k = 0$$

• 
$$y(0) = f(0) * g(0) = f(0)g(0) = 2 \times 1 = 2$$

• k = 1

• 
$$y(1) = f(1) * g(1) = \sum_{i=0}^{1} f(1-i)g(i) = f(1)g(0) + f(0)g(1) = 3 \times 1 + 2 \times 2 = 7$$

• k = 2

• 
$$y(2) = f(2) * g(2) = \sum_{i=0}^{2} f(2-i)g(i) = f(2)g(0) + f(1)g(1) + f(0)g(2) = 4 \times 1 + 3 \times 2 + 2 \times 3 = 16$$

• k = 3

• 
$$y(3) = f(3) * g(3) = \sum_{i=0}^{3} f(3-i)g(i) = f(3)g(0) + f(2)g(1) + f(1)g(2) + f(0)g(3) = 5 \times 1 + 4 \times 2 + 3 \times 1 + 2 \times 0 = 16$$

• k = 4

• 
$$y(4) = f(4) * g(4) = \sum_{i=0}^{4} f(4-i)g(i) = f(3)g(1) + f(2)g(2) = 5 \times 2 + 4 \times 3 = 22$$

• k = 5

• 
$$y(5) = f(5) * g(5) = \sum_{i=0}^{5} f(5-i)g(i) = f(3)g(2) = 5 \times 3 = 15$$

• *k* > 6

• 
$$y(6) = 0$$

## The Convolution Theorem

- The **Convolution** is a fairly complex operation, especially if the output sequence is required over a long time period.
  - Can we avoid this convolution?

#### THE CONVOLUTION THEOREM

• The z-transform of the convolution of two time sequences is equal to the product of their z-transforms.

$$Y(z) = H(z)U(z)$$

- The function H(z) is known as the z-transfer function or simply the **Transfer Function**.
- Applying the convolution allows us to use the z-transform to find the output of a system without using convolution.



## Example

Given the discrete-time system

$$y(k+1) - 0.5y(k) = u(k), y(0) = 0$$

find the impulse response of the system h(k):

- 1. From the difference equation
- 2. Using z-transformation

#### Solution

**1.** Let  $u(k) = \delta(k)$ . Then

k=0 
$$y(1) = 1$$
  
k=1  $y(2) = 0.5y(1) = 0.5$   
k=2  $y(3) = 0.5y(2) = (0.5)^2$   
i.e.,  $h(i) = \begin{cases} (0.5)^{i-1}, & i = 1, 2, 3, ... \\ 0, & i < 1 \end{cases}$ 

**2.** Alternatively, *z*-transforming the difference equation yields the transfer function

$$H(z) = \frac{Y(z)}{U(z)} = \frac{1}{z - 0.5}$$
Time Advance Property:
$$\mathcal{Z}\{f(k+1)\} = zF(z) - zf(0)$$

$$\mathcal{Z}\{f(k+n)\} = z^n F(z) - z^n f(0) - z^{n-1} f(1) - \dots - zf(n-1)$$

Inverse-transforming with the delay theorem gives the impulse response

$$h(i) = \begin{cases} (0.5)^{i-1}, & i = 1, 2, 3, \dots \\ 0, & i < 1 \end{cases}$$

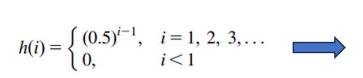
## MATLAB Implementation

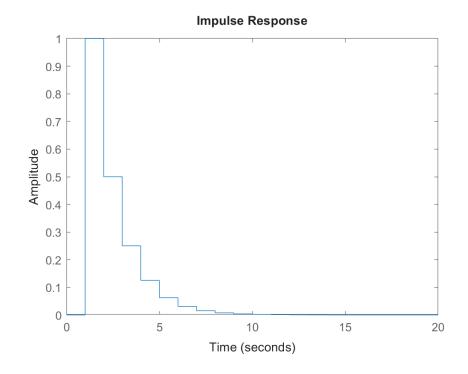
 We can use MATLAB to find and plot the time-domain response of discrete-time systems.

$$H(z) = \frac{1}{z - 0.5}$$

% Define the discrete-time system using transfer function tf % tf(numerator, denominator, sample period) H = tf(1, [1, -0.5], 1);

% Get the system response to an impulse response impulse(H)





## Example 2

Given the discrete-time system

$$y(k + 1) - y(k) = u(k + 1)$$

Find the system transfer function and its response to a sampled unit step

- Solution:
  - The transfer function corresponding to the difference equation is

$$zY(z) - Y(z) = zU(z); H(z) = \frac{Y(z)}{U(z)} = \frac{z}{z-1}$$

• We multiply the transfer function by the sampled unit step's z-transform to obtain

Note: the z-

transform of a unit

step signal is  $\frac{z}{z-1}$ 

$$Y(z) = \left(\frac{z}{z-1}\right) \times \left(\frac{z}{z-1}\right) = \left(\frac{z}{z-1}\right)^2 = z\frac{z}{(z-1)^2}$$

- The z-transform of a unit ramp is  $F(z) = \frac{z}{(z-1)^2}$
- Then, using the time advance property of the z-transform, we have the inverse transform

$$y(i) = \begin{cases} i+1, & i=0, 1, 2, 3, \dots \\ 0, & i < 0 \end{cases}$$

## **Practice Questions**

• Q1: Find the transfer function of the following systems:

$$y(k + 4) + y(k - 1) = u(k)$$

• Solution:

Z-transform 
$$(z^4 - z^{-1})Y(z) = U(z)$$
  $G(z) = \frac{z}{z^5 + 1}$ 

• Q2: Given the discrete-time system

$$y(k+2) - y(k) = 2u(k)$$

- Find the impulse response of the system
- Solution:

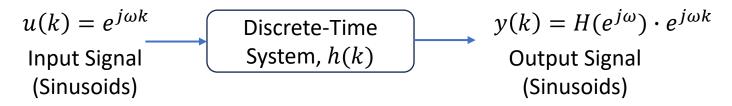
Z-transform: 
$$z^2Y(z) - Y(z) = 2U(z) \rightarrow H(z) = \frac{2}{z^2 - 1} = \frac{1}{z - 1} - \frac{1}{z + 1}$$

 $\frac{1}{z-1}$  can be treated as  $z^{-1}\frac{z}{z-1}$ , and the inverse z-transform is  $1^{k-1}$  for k>1

Thus, the impulse response is 
$$y(x) = \begin{cases} 1^{k-1} - (-1)^{k-1}, & k \ge 1 \\ 0, k < 1 \end{cases}$$

## Frequency Response of Discrete-time Systems

 Frequency Response of discrete-time systems gives the magnitude and phase response of the system to the sinusoids at all frequencies.



Proof:

$$y(k) = h(k) * u(k) = \sum_{i=0}^{k} h(k)u(k-i)$$

• Now,  $u(k) = e^{j\omega k}$ . Thus, we have:

$$y(k) = h(k) * u(k) = \sum_{i=0}^{k} h(k)e^{j\omega(k-i)}$$
$$= \sum_{i=0}^{k} h(k)e^{-j\omega i}e^{j\omega k}$$
$$H(e^{j\omega})$$

## Frequency Response of Discrete-time Systems

$$y(k) = H(e^{j\omega}) \cdot e^{j\omega k}$$

- $H(e^{j\omega})$  is define as the **frequency response** of the discrete-time system.
  - $H(e^{j\omega})$  is a complex number, which can be represented as:

$$H(e^{j\omega}) = |H(e^{j\omega})| \angle H(e^{j\omega})$$

$$y(k) = H(e^{j\omega}) \cdot e^{j(\omega k + \angle H(e^{j\omega}))}$$

$$y(k) = |H(e^{j\omega})| \cdot e^{j(\omega k + \angle H(e^{j\omega}))}$$

Recall Euler equation:

$$e^{j\omega} = \cos(\omega) + j \cdot \sin(\omega)$$

$$y(k) = |H(e^{j\omega})| \cdot \cos\left(\omega k + \angle H(e^{j\omega})\right) + j \cdot |H(e^{j\omega})| \sin\left(\omega k + \angle H(e^{j\omega})\right)$$
If input  $u(t) = \cos(\omega k)$ 
If input  $u(t) = \sin(\omega k)$ 

## Example

Find the steady-state response of the system

$$H(z) = \frac{1}{(z - 0.1)(z - .5)}$$

• due to the sampled sinusoid  $u(k) = 3\cos(0.2 k)$ 

#### Solution:

 Since the input signal is a cosine signal, the output signal should only contain cosine part, i.e.,

$$y(k) = |H(e^{j\omega})| \cdot 3\cos(\omega k + \angle H(e^{j\omega}))$$

$$\omega = 0.2$$

$$y(k) = |H(e^{j0.2})| \cdot 3\cos(0.2k + \angle H(e^{j0.2}))$$

$$= \left|\frac{1}{(e^{j0.2} - 0.1)(e^{j0.2} - 0.5)}\right| 3\cos(0.2k + \angle \frac{1}{(e^{j0.2} - 0.1)(e^{j0.2} - 0.5)})$$

$$= 6.4\cos(0.2k - 0.614)$$

## **Practice Question**

• Find the steady-state response of the systems due to the sinusoidal input  $u(k) = 0.5 \sin(0.4 k)$ 

$$H(z) = \frac{z}{z - 0.4}$$

• Solution:

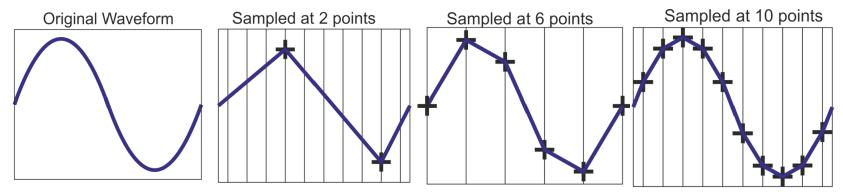
$$H(z) = \frac{z}{z - 0.4} = \frac{1}{1 - 0.4z^{-1}}$$

$$H(e^{j0.4}) = \frac{1}{1 - 0.4e^{-j0.4}} = 1.537 \angle -0.242$$

$$u(k) = 0.5 \times 1.537 \sin(0.4 \, k - 0.242) = 0.769 \sin(0.4k - 0.242)$$

## The Sampling Theorem: Introduction

- Sampling is necessary for the processing of analog data using digital elements.
- Successful digital data processing requires that the samples reflect the Nature of the Analog Signal, and that Analog Signal is Recoverable.
  - The following figures show different discrete-time signals with varying sampling rate.



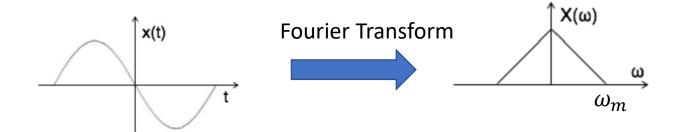
- Obviously, faster sampling would produce distinguishable sequences.
  - Thus, it is obvious that sufficiently fast sampling is a prerequisite for successful digital data processing.
  - How fast is sufficiently fast? -> The **Sampling Theorem** gives **a lower bound** on the **sampling rate** necessary for a given band-limited signal.

# The Sampling Theorem

- The Sampling Theorem:
  - A continuous-time signal can be represented in its samples and can be recovered back when sampling frequency  $f_s$  is **greater than or equal to the twice of the highest frequency** component of the signal:

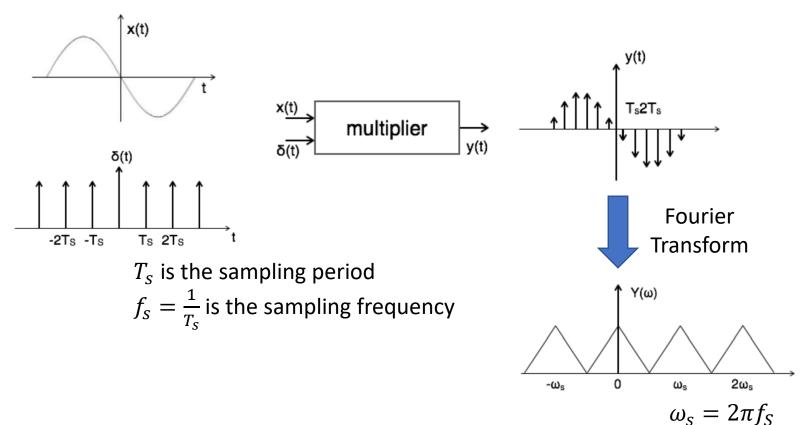
$$f_s \ge 2f_m$$

• Example: consider a continuous time signal x(t). The spectrum (Fourier Transform Result) of x(t) is limited to  $f_m$  Hz (or  $\omega_m$  rad/s).

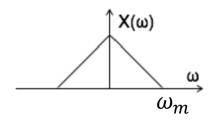


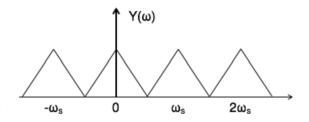
## The Sampling Theorem

• The sampling of x(t) can be obtained by multiplying x(t) with an impulse train  $\delta(t)$ 

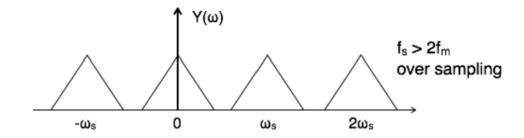


# The Sampling Theorem

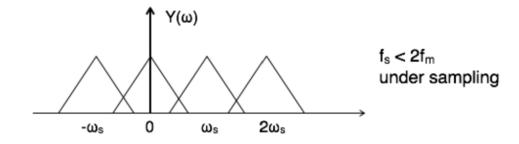




• If  $f_S > 2f_m$ , i.e.,  $\omega_S > 2\omega_m$ 



• If  $f_S < 2f_m$ , i.e.,  $\omega_S < 2\omega_m$ 



# Selection of the Sampling Frequency

- In practice, the sampling rate chosen is often much larger than the lower bound specified in the sampling theorem.
  - The sampling frequency is chosen as  $\omega_s = k\omega_m$ ,  $35 \le k \le 70$

## Example

- Given a first-order system of bandwidth 10 rad/s (i.e.,  $\omega_m=10$  rad/s), select a suitable sampling frequency and find the corresponding sampling period.

  (Any values between 35 to 70 is feasible)
- Solution:
  - A suitable choice of sampling frequency is  $\omega_s=60*\omega_m$  rad/s=60\*10=600 rad/s.
  - The corresponding sampling period is approximately  $T_S = \frac{2\pi}{\omega_S} = \frac{2\pi}{600} = 0.01$  s.

# Selection of the Sampling Frequency: Second-Order Systems

• For a **second-order system**, the bandwidth of the system is approximated by the damped natural frequency

$$\omega_d = \omega_n \sqrt{1 - \zeta^2}$$

• The **sampling frequency** for a second-order system is

$$\omega_s = k\omega_d$$
,  $35 \le k \le 70$ 

# Selection of the Sampling Frequency: Second-Order Systems

#### • Example:

• A second-order closed-loop control system has a damping ratio 0.7 (i.e.,  $\zeta=0.7$ ) and natural frequency 10 rad/s (i.e.,  $\omega_n=10$  rad/s). Select a suitable sampling period for the system.

#### Solution:

Let the sampling frequency be

$$\omega_s \ge 35\omega_d$$

$$= 35\omega_n \sqrt{1 - \zeta^2}$$

$$= 350\sqrt{1 - 0.49}$$

$$= 249.95 \text{ rad/s}$$

The corresponding sampling period is  $T = 2\pi/\omega_s \le 0.025$  s.

## **Practice Questions**

## Questions:

- For a first-order system, the system bandwidth is 20 rad/s. Select the suitable sampling period.
- For a second-order system, the system natural frequency is 5 rad/s and the damping ratio is 0.7. Find the suitable sampling period.

#### Solution:

- $\omega_m=20$  rad/s, we can choose k=40, thus,  $\omega_s=800$  rad/s and  $T=\frac{2\pi}{\omega_s}=0.00785$  s =7.85ms
- For second order systems, the system bandwidth is  $\omega_d = \omega_n \sqrt{1-\zeta^2}$ 
  - Thus,  $\omega_d = 5\sqrt{1 0.7^2} = 3.57 \text{ rad/s}$
  - We can choose k = 70, thus

$$T = \frac{2\pi}{\omega_s} = \frac{2\pi}{70\omega_d} = 0.025s$$
 Let  $T = 25$  ms.