

# **ELG3125 Signal and System Analysis**

## **LTI Systems and Convolution Sum**

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**Slides and assignments:**

**<http://www.site.uottawa.ca/~msade033/signal/>**

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# Outline

1. Differential equations
2. Difference equations
3. Convolution sums

\*To learn to simulate LTI systems in continuous time and discrete time, represented by differential equations, difference equations, or impulse response.

# Continuous-Time LTI System

- $N$ -th order linear constant-coefficient differential equation:

$$\sum_{k=0}^N a_k \frac{d^k y(t)}{dt^k} = \sum_{k=0}^M b_k \frac{d^k x(t)}{dt^k}$$

- $N$ -th order linear constant-coefficient differential equation:

$$A = [a_N, a_{N-1}, \dots, a_0], \quad B = [b_M, b_{M-1}, \dots, b_0]$$

- To find the impulse response  $h(t)$  of the system:

$$\text{impulse}(B, A, t)$$

- To compute the output given the input  $x(t)$ :

$$\text{lsim}(B, A, x, t)$$

## Example: Continuous-Time LTI System

- Second order LTI system

$$\frac{d^2 y(t)}{dt^2} + 2 \frac{dy(t)}{dt} + 3y(t) = \frac{dx(t)}{dt} + 6x(t)$$

$$A = [1 \ 2 \ 3] \quad B = [0 \ 1 \ 6]$$

- Coefficients are highest-order first from left to right in the vector.
- The size of A and B should be the same.

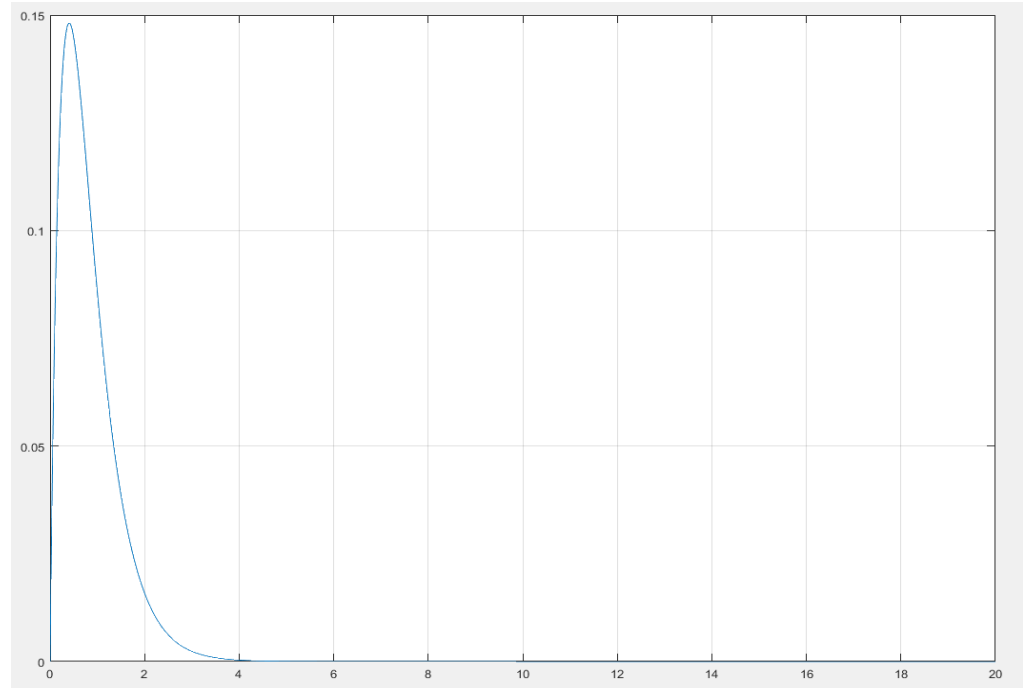
# Assignment – Question 1

1. For the following continuous time LTI system:

$$\frac{d^2 y(t)}{dt^2} + 5 \frac{dy(t)}{dt} + 6y(t) = x(t) \quad (\text{system initially at rest})$$

a) Find and plot using Matlab the impulse response  $h(t)$  of the system, using the function *impulse(b,a,t)* where  $b$  and  $a$  correspond to the coefficients in the differential equation (i.e.  $a=[1, 5, 6]$  and  $b=[0, 0, 1]$ ).

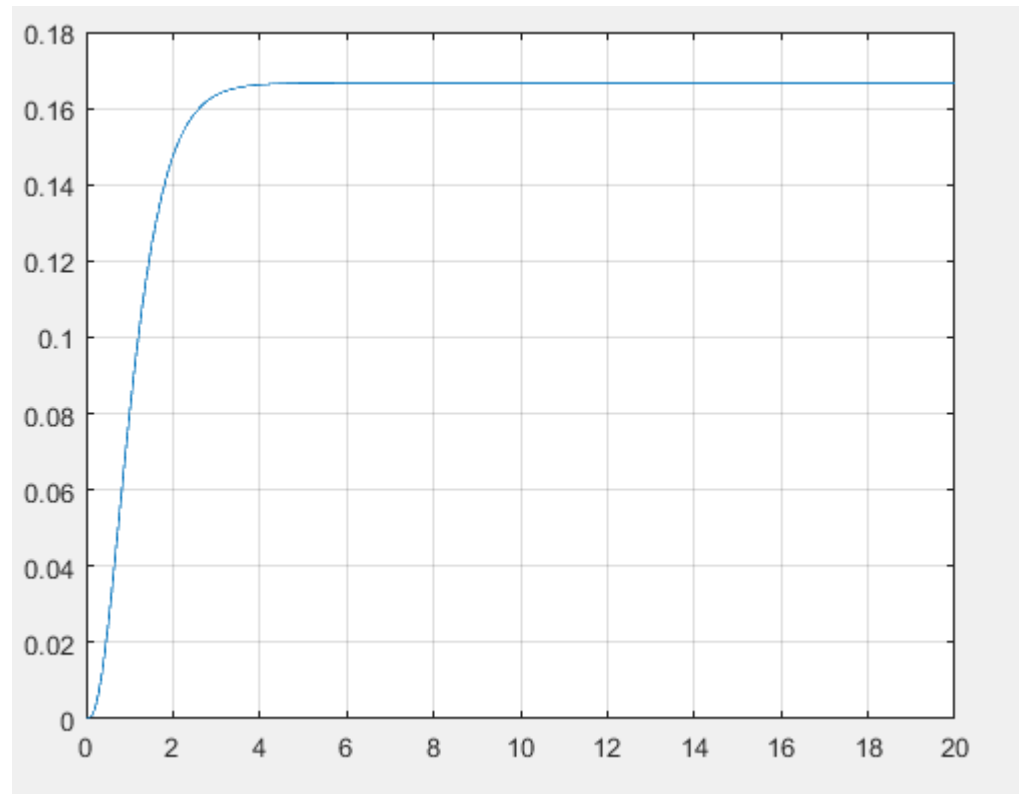
```
a = [1 5 6];  
b = [0 0 1];  
impulse(b, a, t);  
grid on;
```



# Assignment ...

b). For an input signal defined as:  $x(t) = (1 - e^{-3t})u(t)$   
compute the output  $y(t)$  of that system, using the function *lsim(b,a,x,t)*.  
Use a time range long enough to see the  $y(t)$  signal stabilize.

```
a=[1, 5, 6];  
b=[0, 0, 1];  
x=1-exp(-3*t);  
c=lsim(b,a,x,t);  
plot(t,c);  
grid on;
```



# Discrete-Time LTI System

- $N$ th-order linear constant-coefficient difference equation

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

- Describe the system in Matlab

$$A = [a_0, a_1, \dots, a_N], B = [b_0, b_1, \dots, b_M]$$

- To find the impulse response  $h[n]$  of the system:

$$\text{impz}(B,A,n)$$

- To compute the output given the input  $x[n]$ :

$$\text{filter}(B,A,x)$$

# Example: Discrete-Time LTI System

- 3rd order LTI system

$$2y[n] + 5y[n-1] + y[n-2] + 8y[n-3] = x[n]$$

- Coefficients are most present time first from left to right in the vector

$$A=[2,5,1,8] \quad B=[1,0,0,0]=[1,\text{zeros}(1,3)]$$



# Convolution sum

- Mathematical formula

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

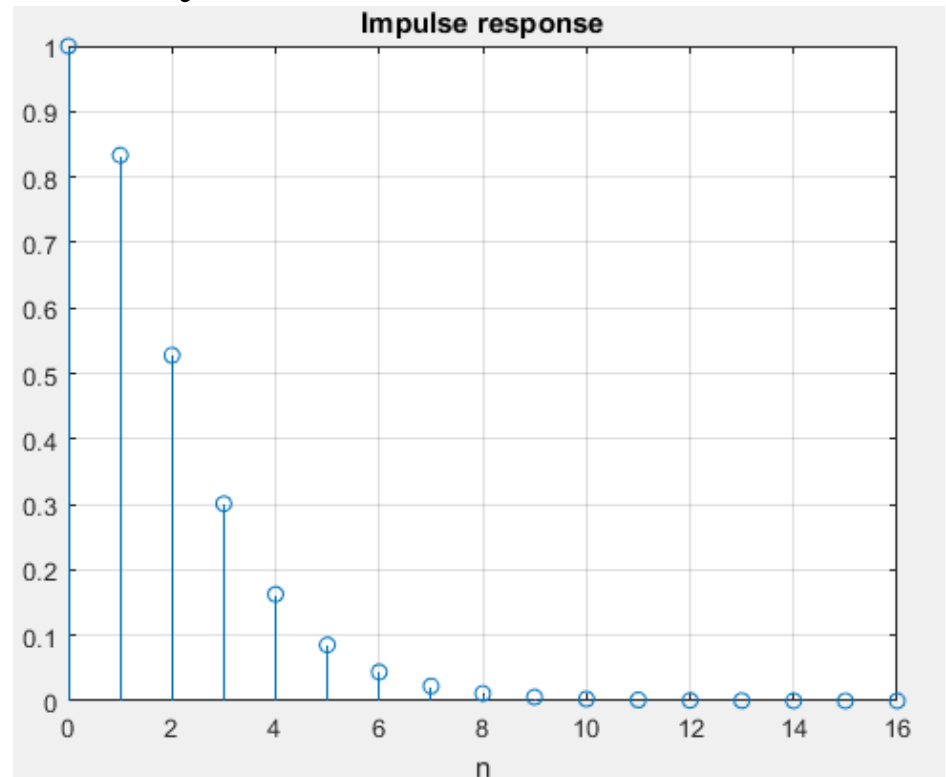
- Matlab function: `y=conv(x,h)`
- Length of signal `x[n]`: `length(x)`

## Assignment – Question 2

2. For the following discrete time LTI system:

$$y[n] - \frac{5}{6}y[n-1] + \frac{1}{6}y[n-2] = x[n] \quad (\text{system initially at rest})$$

```
a=[1, -5/6, 1/6];  
b=[1 0 0];  
n=0:16;  
h=impz(b,a,n);  
figure(1);  
stem(n,h),grid  
xlabel('n');  
ylabel('h[n]');  
title('Impulse response');
```



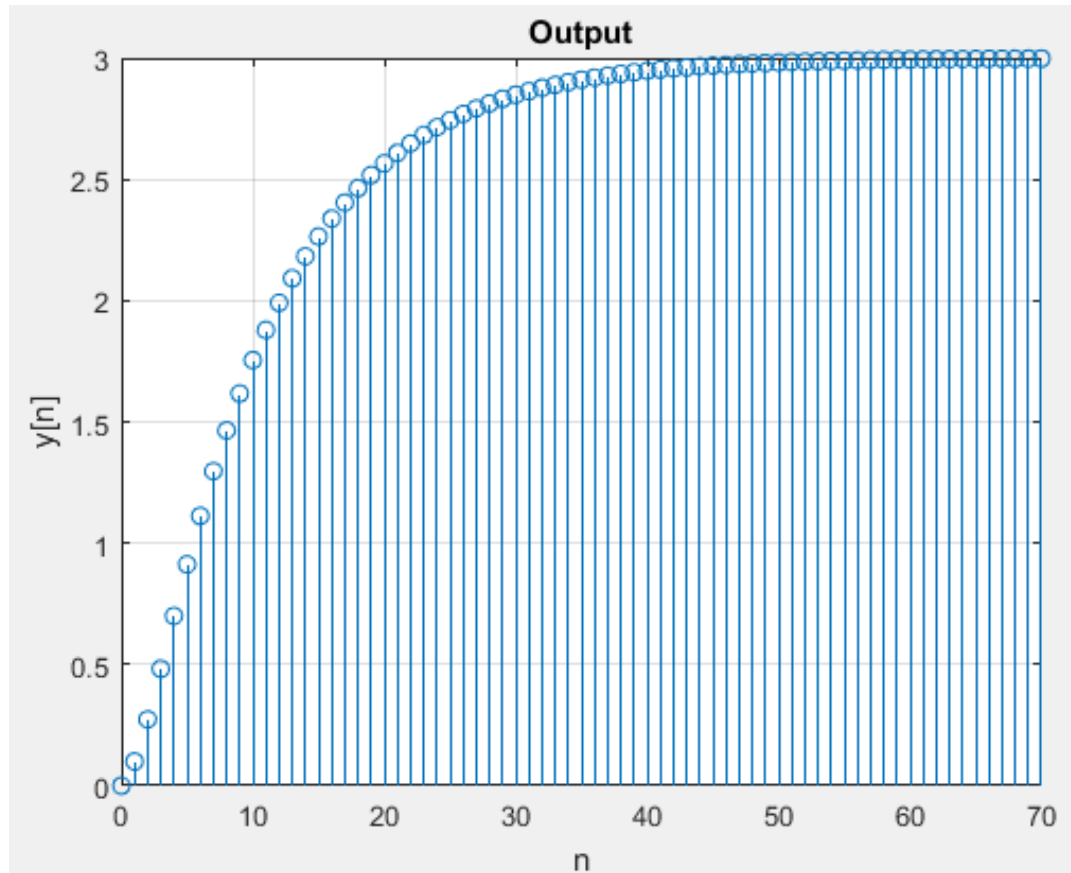
# Assignment ...

b) For an input signal defined as:  $x[n] = (1 - 0.9^n)u[n]$

compute the output  $y[n]$  of that system, using the function `filter`.

Use a time range long enough to see the  $y[n]$  signal stabilize.

```
a = [1, -5/6, 1/6];  
b = [1 0 0];  
n = 0:70;  
x = 1 - power(0.9, n);  
y = filter(b, a, x);  
stem(n, y), grid;  
xlabel('n');  
ylabel('y[n]');  
title('Output');
```



# Learn How to Operate Music Files

- Read a music file in Matlab: *audioread()*
- Listen to the music: *Sound()*

## Assignment – Question 3

- First obtain a discrete time input signal  $x[n]$  by reading the file "Audio1.wav" from the course's website. This signal has a duration (length) of 190912 samples, with a sampling frequency of 16000 samples/sec. Visualize the resulting signal and listen to it.
- The page:  
<http://www.site.uottawa.ca/~msade033/signal/>

```
x=audioread('Audio1.wav');  
sound(x); % Please don't execute the  
syntax unless you have headphones!
```

## Assignment ...

$$h[n] = 0.1 \times (0.99)^n, 0 \leq n \leq 1000$$

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

c) Use the function `conv( )` to compute  $y[n]$ . Observe the size of the result  $y[n]$ , plot  $y[n]$  and listen to it. What do you observe when directly comparing with  $x[n]$ ? Does it sound different ?

```
x=audioread('Audio1.wav');  
n=0:1000;  
h=0.1*(0.99).^n;  
y=conv(x,h);  
stem(y);  
length(y) % has to be equal to  
length(x)+length(h)-1
```

## Assignment ...

- d) Now use the function `filter` to compute  $y[n]$ , where the parameter `b` in the function `filter` corresponds to  $h[n]$  and the parameter `a` is set to 1.0. Like the function `conv( )`, the function `filter` also computes the result of a discrete time convolution in this case, but not over the same interval of output  $y[n]$  values. Explain the difference.

```
x=audioread('Audio1.wav');  
n=0:1000;  
h=0.1*(0.99).^n;  
y=filter(h,1,x);  
stem(y);  
length(y) % has to be equal to length(x)
```

# Thank you!