# 6.3000: Signal Processing

## **Convolution and Filtering**

#### time domain

#### frequency domain

$$y(t) = (h * x)(t) = \int h(\tau)x(t - \tau) \, d\tau$$

$$Y(\omega) = H(\omega)X(\omega)$$

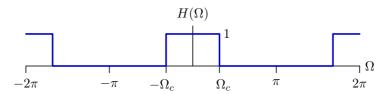
$$y[n] = (h * x)[n] = \sum_{m} h[m]x[n-m]$$

$$Y(\Omega) = H(\Omega)X(\Omega)$$

March 13, 2025

#### The "Ideal" Low-Pass Filter

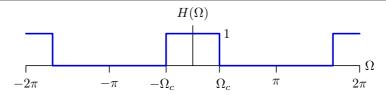
Consider a system characterized by the following purely real frequency response:



Such a system is called a **low-pass filter**, because it allows low frequencies to pass through unmodified, while attenuating high frequencies.

We could apply this filter to a signal by multiplying the DTFT of that signal by the values above. But we could also apply the filter by operating in the time domain.

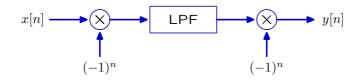
#### The "Ideal" Low-Pass Filter



We can apply this filter to a signal by convolving with its unit sample response. What is the unit sample response of the system whose frequency response is shown above?

$$h[n] = \frac{1}{2\pi} \int_{2\pi} H(\Omega) e^{j\Omega n} d\Omega$$
$$= \frac{1}{2\pi} \int_{-\Omega_c}^{\Omega_c} e^{j\Omega n} d\Omega$$
$$= \frac{1}{2\pi} \frac{1}{jn} e^{j\Omega n} \Big|_{\Omega = -\Omega_c}^{\Omega_c}$$
$$= \frac{\sin(\Omega_c n)}{\pi n}$$

Consider the following system, where LPF represents a lowpass filter of the form discussed on the previous slides.



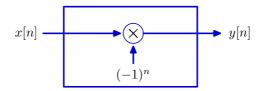
How many of the following statements are true?

- The transformation from x[n] to y[n] is linear.
- The transformation from x[n] to y[n] is time invariant.
- The transformation from x[n] to y[n] is a high-pass filter.

# **Consider Each Part Separately**

Start with the multiplier.

Is the following system linear?



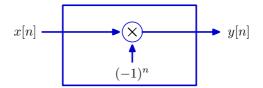
Assume the response to the input  $x_1[n]$  is  $y_1[n]$  and the response to the input  $x_2[n]$  is  $y_2[n]$ . Calculate the response to  $x[n] = \alpha x_1[n] + \beta x_2[n]$ .

$$y[n] = (-1)^n x[n]$$
  
=  $(-1)^n (\alpha x_1[n] + \beta x_2[n])$   
=  $\alpha (-1)^n x_1[n] + \beta (-1)^n x_2[n]$   
=  $\alpha y_1[n] + \beta y_2[n]$ 

∴ linear!

## **Consider Each Part Separately**

Is this system time-invariant?



Assume the response to the input  $x_1[n]$  is  $y_1[n]$ . Calculate the response to  $x[n] = x_1[n - n_0]$ .

$$y[n] = (-1)^n x[n]$$
  
=  $(-1)^n x_1[n - n_0]$   
=  $(-1)^{n_0} (-1)^{n - n_0} x_1[n - n_0]$   
=  $(-1)^{n_0} y_1[n - n_0]$ 

Shifting the input by  $n_0$  samples shifts does not just shift the output by  $n_0$  samples.

∴ this system is not time-invariant!

# **Consider Each Part Separately**

Is the LPF linear? time-invariant?



The defining property of a **filter** is that its output is a weighted sum of (possibly) shifted versions of the frequencies in the input, so that

$$Y(\Omega) = H(\Omega)X(\Omega) \,.$$

Multiplication in frequency is the same as convolution in time, so

$$y[n] = (x * h_L)[n]$$

where  $h_L[\cdot]$  represents the unit-sample response of the filter.

Convolution is both linear and time invariant (as shown on the next slides).

#### **Convolution is Linear**

Assume the response to the input  $x_1[n]$  is  $y_1[n]$  and the response to the input  $x_2[n]$  is  $y_2[n]$ . Calculate the response to  $x[n] = \alpha x_1[n] + \beta x_2[n]$ .

$$y[n] = \sum_{m=-\infty}^{\infty} x[m]h[n-m]$$
  
= 
$$\sum_{m=-\infty}^{\infty} \left(\alpha x_1[m] + \beta x_2[m]\right)h[n-m]$$
  
= 
$$\alpha \sum_{m=-\infty}^{\infty} x_1[m]h[n-m] + \beta \sum_{m=-\infty}^{\infty} x_2[m]h[n-m]$$
  
= 
$$\alpha y_1[n] + \beta y_2[n]$$

∴ linear!

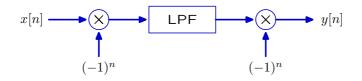
#### **Convolution is Time-Invariant**

Assume the response to the input  $x_1[n]$  is  $y_1[n] = (x_1 * h)[n]$ . Calculate the response to  $x[n] = x_1[n-n_0]$ .

$$\begin{split} y[n] &= (x * h)[n] \\ &= \sum_{m=-\infty}^{\infty} x[m]h[n-m] \\ &= \sum_{m=-\infty}^{\infty} x_1[m-n_0]h[n-m] \\ &= \sum_{l=-\infty}^{\infty} x_1[l]h[n-(n_0+l)] \\ &= \sum_{l=-\infty}^{\infty} x_1[l]h[(n-n_0)-l] \\ &= (x_1 * h)[n-n_0] = y_1[n-n_0] \end{split}$$

∴ time-invariant!

This system is the cascade of three linear subsystems, two of which are time varying.



Thus we know that the composite system is linear.

- The transformation from x[n] to y[n] is linear.  $\sqrt{}$
- The transformation from x[n] to y[n] is time invariant.
- The transformation from x[n] to y[n] is a high-pass filter.

The first is true. Not sure about the others.

Could the cascade of two time-varying systems be time-invariant?

Yes. Think about the cascade of  $\times (-1)^n$  with  $\times (-1)^n$ .

w

z

y

Determine an expression for y[n] in terms of x[n] using a time-domain approach. Assume that the unit sample response of the LPF is  $h_L[n]$ .

$$x[n] \xrightarrow{w[n]} LPF \xrightarrow{z[n]} \xrightarrow{y[n]} y[n]$$

$$(-1)^{n} (-1)^{n} (-1)^{n}$$

$$[n] = (-1)^{n}x[n]$$

$$[n] = (w * h_{L})[n] = \sum_{m=-\infty}^{\infty} (-1)^{m}x[m]h_{L}[n-m]$$

$$[n] = (-1)^{n}z[n] = (-1)^{n}\sum_{m=-\infty}^{\infty} (-1)^{m}x[m]h_{L}[n-m]$$

$$= \sum_{m=-\infty}^{\infty} (-1)^{n+m}x[m]h_{L}[n-m] = \sum_{m=-\infty}^{\infty} (-1)^{n-m}x[m]h_{L}[n-m]$$

$$= (x * h_{H})[n] \text{ where } h_{H}[n] = (-1)^{n}h_{L}[n]$$

Find an expression for the unit sample response of a lowpass filter that passes frequencies in the range  $-\Omega_c < \Omega < \Omega_c$ .

$$H_{L}(\Omega)$$

$$H_{L}(\Omega)$$

$$H_{L}[n] = \frac{1}{2\pi} \int_{2\pi} H_{L}(\Omega) e^{j\Omega n} d\Omega$$

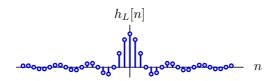
$$= \frac{1}{2\pi} \int_{-\Omega_{c}}^{\Omega_{c}} e^{j\Omega n} d\Omega$$

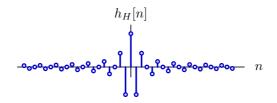
$$= \frac{1}{2\pi} \left[ \frac{e^{j\Omega n}}{jn} \right]_{-\Omega_{c}}^{\Omega_{c}}$$

$$= \frac{\sin(\Omega_{c} n)}{\pi n}$$

:  $h_H[n] = (-1)^n h_L[n] = (-1)^n \frac{\sin(\Omega_c n)}{\pi n}$ 

Plot  $h_L[n]$  and  $h_H[n]$  for  $\Omega_c = \frac{\pi}{3}$ .





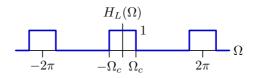
Determine an expression for  $H_H(\Omega)$ .

$$H_H(\Omega) = \sum_{n=-\infty}^{\infty} h_H[n] e^{-j\Omega n}$$
$$= \sum_{n=-\infty}^{\infty} (-1)^n h_L[n] e^{-j\Omega n}$$
$$= \sum_{n=-\infty}^{\infty} e^{j\pi n} h_L[n] e^{-j\Omega n}$$
$$= \sum_{n=-\infty}^{\infty} h_L[n] e^{-j(\Omega - \pi)n}$$
$$= H_L(\Omega - \pi)$$

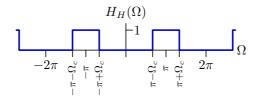
This is an example of the frequency shifting property:

If 
$$x[n] \stackrel{\text{DTFT}}{\Rightarrow} X(\Omega)$$
 then  $e^{j\Omega_o n} x[n] \stackrel{\text{DTFT}}{\Rightarrow} X(\Omega - \Omega_o).$ 

Plot  $H_H(\Omega)$ . Compare to  $H_L(\Omega)$ .

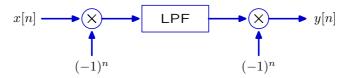


 $H_H(\Omega) = H_L(\Omega - \pi)$ 



 $H_H(\Omega)$  is a highpass filter!

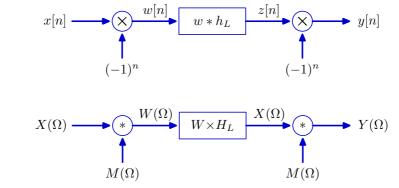
We have just shown that the original system is a highpass system.



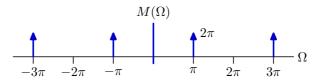
Therefore all of the following are true!

- The transformation from x[n] to y[n] is linear.  $\sqrt{}$
- The transformation from x[n] to y[n] is time invariant.  $\sqrt{}$
- The transformation from x[n] to y[n] is a high-pass filter.  $\sqrt{}$

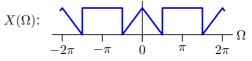
Alternatively, we could solve this problem in the frequency domain.



Since  $(-1)^n = e^{j\pi n}$ ,  $M(\Omega)$  is a complex sinusoid with frequency  $\pi$ .



Assume  $X(\Omega)$  differs at high and low frequencies, as shown below.

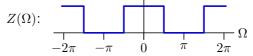


Convolving with  $M(\Omega)$  shifts  $X(\Omega)$  by  $\Omega=\pi$ :

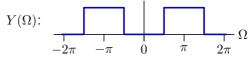
I

$$W(\Omega)$$
:  $-2\pi$   $-\pi$   $0$   $\pi$   $2\pi$   $\Omega$ 

Low pass filtering removes the high frequencies. Assume  $\Omega_c = \pi/2$ :



Convolving with  $M(\Omega)$  shifts the result by  $\Omega = \pi$ :



 $Y(\Omega)$  is a highpass version of  $X(\Omega)$ .