

# Simple Digital Filters

- Later in the course we shall review various methods of designing frequency-selective filters satisfying prescribed specifications
- We now describe several low-order FIR and IIR digital filters with reasonable selective frequency responses that often are satisfactory in a number of applications

# Simple FIR Digital Filters

- FIR digital filters considered here have integer-valued impulse response coefficients
- These filters are employed in a number of practical applications, primarily because of their simplicity, which makes them amenable to inexpensive hardware implementations

# Simple FIR Digital Filters

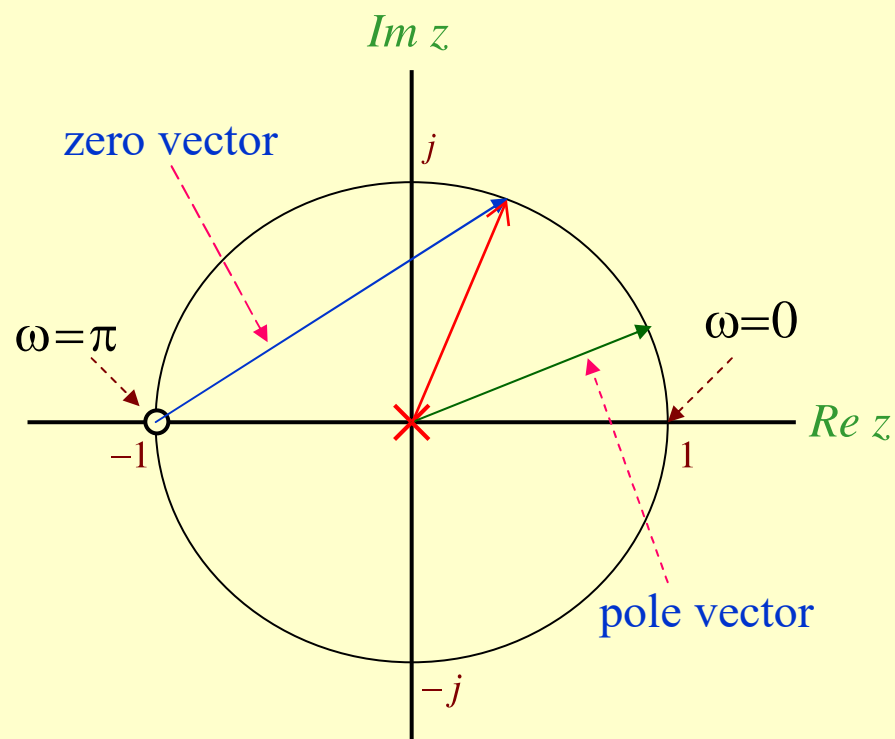
## Lowpass FIR Digital Filters

- The simplest lowpass FIR digital filter is the 2-point moving-average filter given by

$$H_0(z) = \frac{1}{2}(1 + z^{-1}) = \frac{z + 1}{2z}$$

- The above transfer function has a zero at  $z = -1$  and a pole at  $z = 0$
- Note that here the pole vector has a unity magnitude for all values of  $\omega$

# Simple FIR Digital Filters



# Simple FIR Digital Filters

- On the other hand, as  $\omega$  increases from 0 to  $\pi$ , the magnitude of the zero vector decreases from a value of 2, the diameter of the unit circle, to 0
- Hence, the magnitude response  $|H_0(e^{j\omega})|$  is a monotonically decreasing function of  $\omega$  from  $\omega = 0$  to  $\omega = \pi$

# Simple FIR Digital Filters

- The maximum value of the magnitude function is 1 at  $\omega = 0$ , and the minimum value is 0 at  $\omega = \pi$ , i.e.,

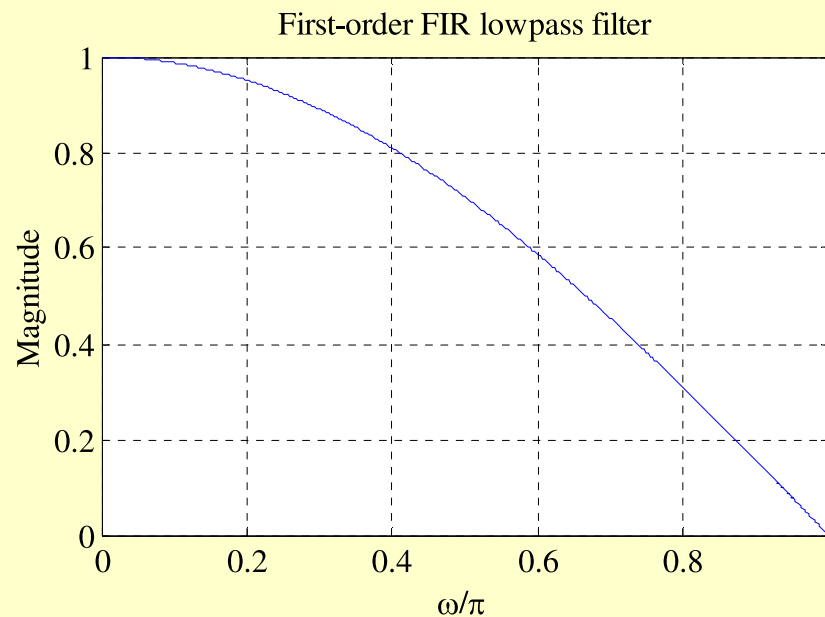
$$|H_0(e^{j0})| = 1, \quad |H_0(e^{j\pi})| = 0$$

- The frequency response of the above filter is given by

$$H_0(e^{j\omega}) = e^{-j\omega/2} \cos(\omega/2)$$

# Simple FIR Digital Filters

- The magnitude response  $|H_0(e^{j\omega})| = \cos(\omega/2)$  can be seen to be a monotonically decreasing function of  $\omega$



# Simple FIR Digital Filters

- The frequency  $\omega = \omega_c$  at which

$$\left| H_0(e^{j\omega_c}) \right| = \frac{1}{\sqrt{2}} \left| H_0(e^{j0}) \right|$$

is of practical interest since here the gain  $G(\omega_c)$  in dB is given by

$$\begin{aligned} G(\omega_c) &= 20 \log_{10} \left| H(e^{j\omega_c}) \right| \\ &= 20 \log_{10} \left| H(e^{j0}) \right| - 20 \log_{10} \sqrt{2} \cong -3 \text{ dB} \end{aligned}$$

since the dc gain  $G(0) = 20 \log_{10} \left| H(e^{j0}) \right| = 0$



# Simple FIR Digital Filters

- Thus, the gain  $G(\omega)$  at  $\omega = \omega_c$  is approximately 3 dB less than the gain at  $\omega = 0$
- As a result,  $\omega_c$  is called the **3-dB cutoff frequency**
- To determine the value of  $\omega_c$  we set
$$|H_0(e^{j\omega_c})|^2 = \cos^2(\omega_c / 2) = \frac{1}{2}$$
which yields  $\omega_c = \pi / 2$

# Simple FIR Digital Filters

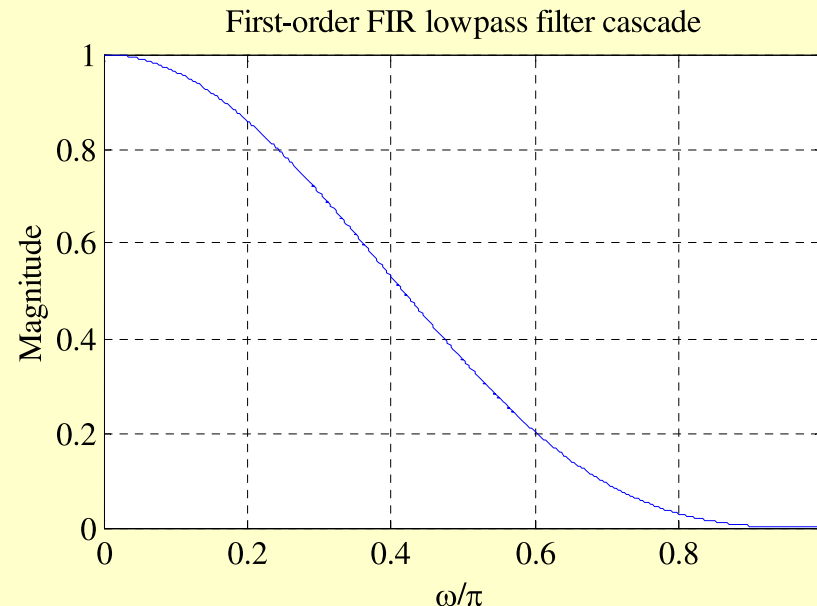
- The 3-dB cutoff frequency  $\omega_c$  can be considered as the passband edge frequency
- As a result, for the filter  $H_0(z)$  the passband width is approximately  $\pi/2$
- The stopband is from  $\pi/2$  to  $\pi$
- Note:  $H_0(z)$  has a zero at  $z = -1$  or  $\omega = \pi$ , which is in the stopband of the filter

# Simple FIR Digital Filters

- A cascade of the simple FIR filter

$$H_0(z) = \frac{1}{2}(1 + z^{-1})$$

results in an improved lowpass frequency response as illustrated below for a cascade of 3 sections



What are the coeffs. of the impulse response of this filter ?

# Simple FIR Digital Filters

- The 3-dB cutoff frequency of a cascade of  $M$  sections is given by

$$\omega_c = 2 \cos^{-1}(2^{-1/2M})$$

- For  $M = 3$ , the above yields  $\omega_c = 0.302\pi$
- Thus, the cascade of first-order sections yields a sharper magnitude response but at the expense of a decrease in the width of the passband

# Simple FIR Digital Filters

- A better approximation to the ideal lowpass filter is given by a higher-order moving-average filter

Compare the two filters using Matlab

- Signals with rapid fluctuations in sample values are generally associated with high-frequency components
- These high-frequency components are essentially removed by an moving-average filter resulting in a smoother output waveform

# Simple FIR Digital Filters

## Highpass FIR Digital Filters

- The simplest highpass FIR filter is obtained from the simplest lowpass FIR filter by

replacing  $z$  with  $-z$

i.e.:  $\exp(j\omega) \Rightarrow$

$-\exp(j\omega) = \exp(j(\omega + \pi))$

- This results in

$$H_1(z) = \frac{1}{2}(1 - z^{-1})$$

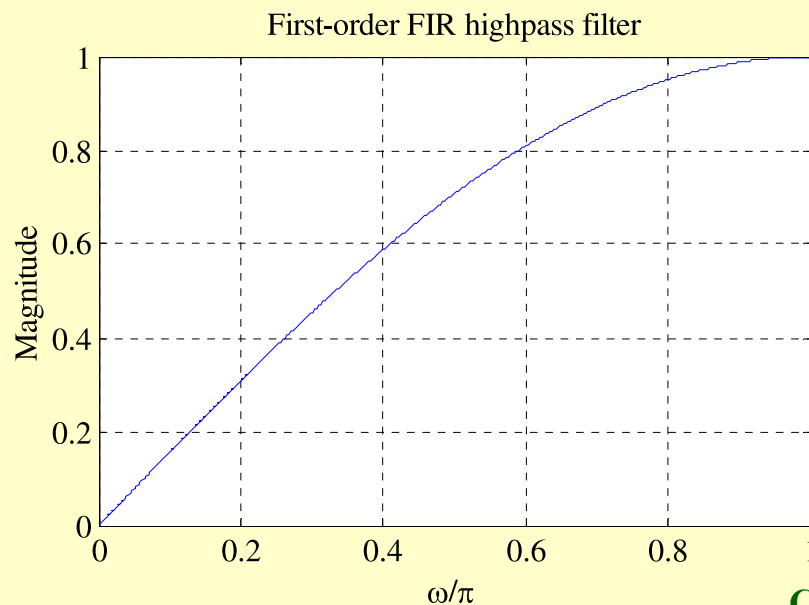
First derivative filter (backward)

# Simple FIR Digital Filters

- Corresponding frequency response is given by

$$H_1(e^{j\omega}) = j e^{-j\omega/2} \sin(\omega/2)$$

whose magnitude response is plotted below



# Simple FIR Digital Filters

- The monotonically increasing behavior of the magnitude function can again be demonstrated by examining the pole-zero pattern of the transfer function  $H_1(z)$
- The highpass transfer function  $H_1(z)$  has a zero at  $z = 1$  or  $\omega = 0$  which is in the stopband of the filter



# Simple FIR Digital Filters

- Improved highpass magnitude response can again be obtained by cascading several sections of the first-order highpass filter
- Alternately, a higher-order highpass filter of the form

$$H_1(z) = \frac{1}{M} \sum_{n=0}^{M-1} (-1)^n z^{-n}$$

is obtained by replacing  $z$  with  $-z$  in the transfer function of a moving average filter

# Simple FIR Digital Filters

- An application of the FIR highpass filters is in moving-target-indicator (MTI) radars
- In these radars, interfering signals, called **clutters**, are generated from fixed objects in the path of the radar beam
- The clutter, generated mainly from ground echoes and weather returns, has frequency components near zero frequency (dc)

# Simple FIR Digital Filters

- The clutter can be removed by filtering the radar return signal through a **two-pulse canceler**, which is the first-order FIR highpass filter  $H_1(z) = \frac{1}{2}(1 - z^{-1})$
- For a more effective removal it may be necessary to use a **three-pulse canceler** obtained by cascading two two-pulse cancelers

I.e.: derivative of the derivative  
--> second derivative filter

# Simple IIR Digital Filters

## Lowpass IIR Digital Filters

- We have shown earlier that the first-order causal IIR transfer function

$$H(z) = \frac{K}{1 - \alpha z^{-1}}, \quad 0 < \alpha < 1$$

has a lowpass magnitude response for  $\alpha > 0$

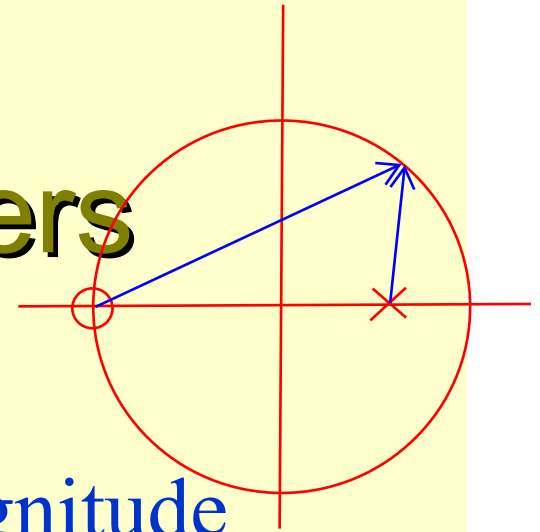
# Simple IIR Digital Filters

- An improved **lowpass magnitude response** is obtained by adding a factor  $(1 + z^{-1})$  to the numerator of transfer function

$$H(z) = \frac{K(1 + z^{-1})}{1 - \alpha z^{-1}}, \quad 0 < \alpha < 1$$

- This forces the magnitude response to have a zero at  $\omega = \pi$  in the **stopband** of the filter

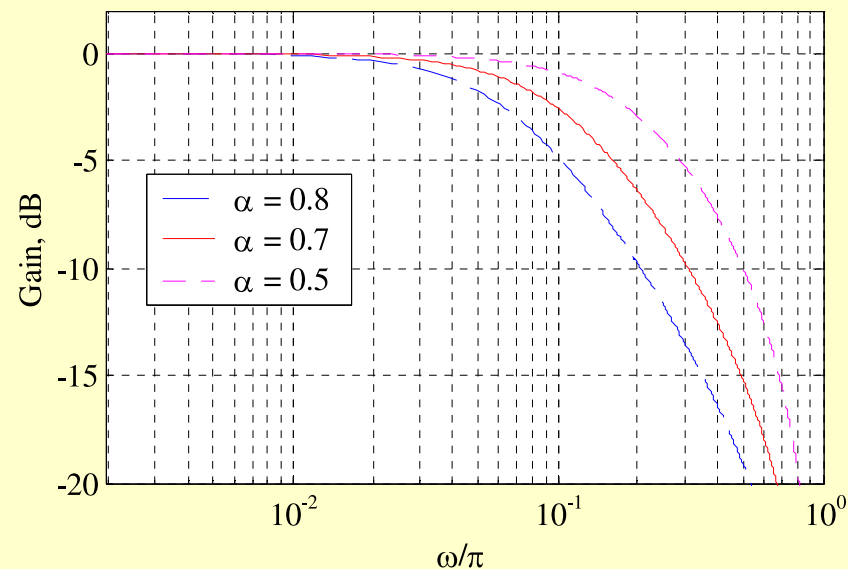
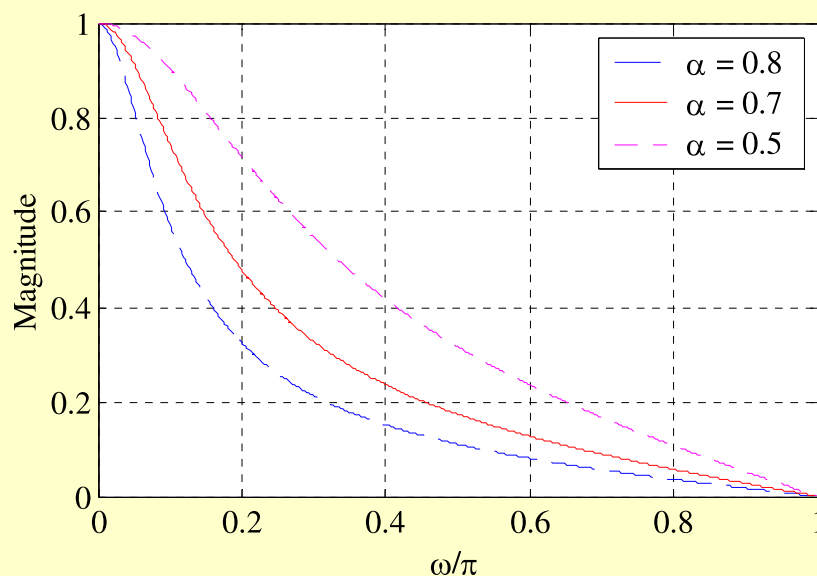
# Simple IIR Digital Filters



- $H_{LP}(z)$  has a real pole at  $z = \alpha$
- As  $\omega$  increases from 0 to  $\pi$ , the magnitude of the zero vector decreases from a value of 2 to 0, whereas, for a positive value of  $\alpha$ , the magnitude of the pole vector increases from a value of  $1 - \alpha$  to  $1 + \alpha$
- The maximum value of the magnitude function is 1 at  $\omega = 0$ , and the minimum value is 0 at  $\omega = \pi$

# Simple IIR Digital Filters

- i.e.,  $|H_{LP}(e^{j0})| = 1$ ,  $|H_{LP}(e^{j\pi})| = 0$
- Therefore,  $|H_{LP}(e^{j\omega})|$  is a monotonically decreasing function of  $\omega$  from  $\omega = 0$  to  $\omega = \pi$  as indicated below



# Simple IIR Digital Filters

- The squared magnitude function is given by

$$|H_{LP}(e^{j\omega})|^2 = \frac{(1-\alpha)^2(1+\cos\omega)}{2(1+\alpha^2-2\alpha\cos\omega)}$$

- The derivative of  $|H_{LP}(e^{j\omega})|^2$  with respect to  $\omega$  is given by

$$\frac{d |H_{LP}(e^{j\omega})|^2}{d\omega} = \frac{-(1-\alpha)^2(1+2\alpha+\alpha^2)\sin\omega}{2(1-2\alpha\cos\omega+\alpha^2)^2}$$



# Simple IIR Digital Filters

$d |H_{LP}(e^{j\omega})|^2 / d\omega \leq 0$  in the range  $0 \leq \omega \leq \pi$   
verifying again the monotonically decreasing  
behavior of the magnitude function

- To determine the 3-dB cutoff frequency  
we set

$$|H_{LP}(e^{j\omega_c})|^2 = \frac{1}{2}$$

in the expression for the square magnitude  
function resulting in

# Simple IIR Digital Filters

$$\frac{(1-\alpha)^2(1+\cos\omega_c)}{2(1+\alpha^2-2\alpha\cos\omega_c)} = \frac{1}{2}$$

or

$$(1-\alpha)^2(1+\cos\omega_c) = 1 + \alpha^2 - 2\alpha\cos\omega_c$$

which when solved yields

$$\cos\omega_c = \frac{2\alpha}{1+\alpha^2}$$

- The above quadratic equation can be solved for  $\alpha$  yielding two solutions

# Simple IIR Digital Filters

- The solution resulting in a stable transfer function  $H_{LP}(z)$  is given by

$$\alpha = \frac{1 - \sin \omega_c}{\cos \omega_c}$$

# Simple IIR Digital Filters

## Highpass IIR Digital Filters

- A first-order causal highpass IIR digital filter has a transfer function given by

$$H_{HP}(z) = \frac{1 + \alpha}{2} \left( \frac{1 - z^{-1}}{1 - \alpha z^{-1}} \right)$$

from the LP filter:  
 $z \Leftrightarrow -z$  and  
redefine  $\alpha$

where  $|\alpha| < 1$  for stability

- The above transfer function has a zero at  $z = 1$  i.e., at  $\omega = 0$  which is in the stopband

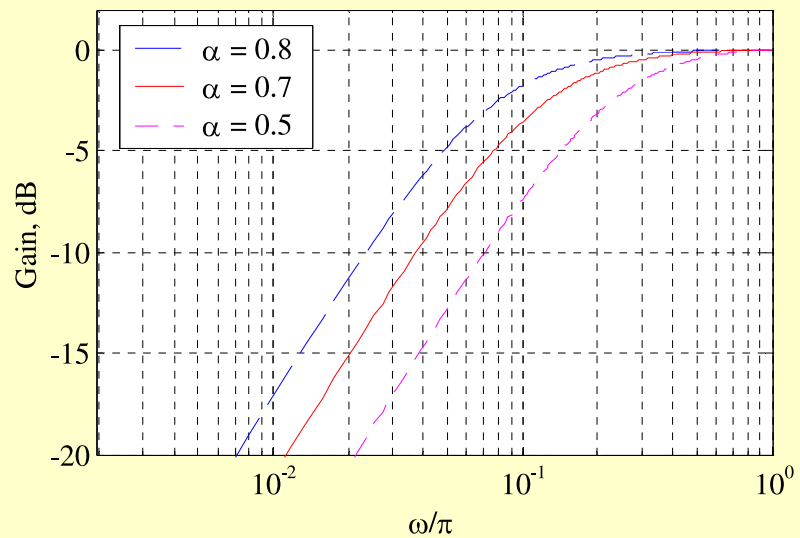
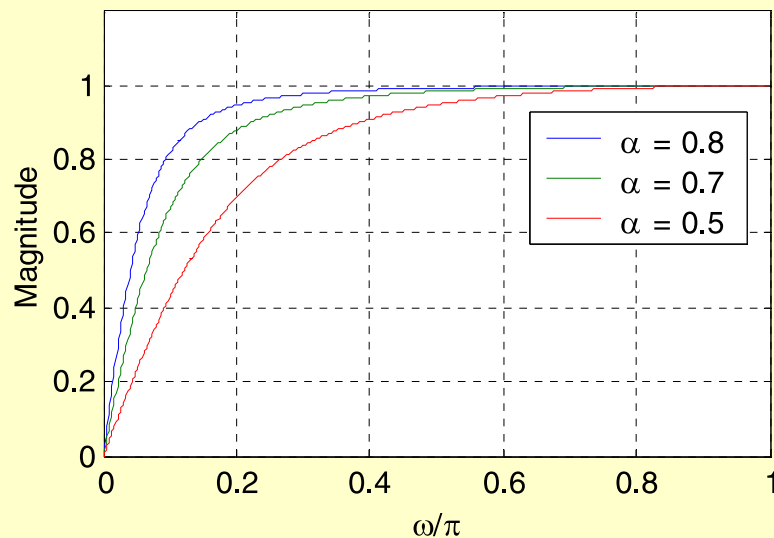
# Simple IIR Digital Filters

- Its 3-dB cutoff frequency  $\omega_c$  is given by

$$\alpha = (1 - \sin \omega_c) / \cos \omega_c$$

which is the same as that of  $H_{LP}(z)$

- Magnitude and gain responses of  $H_{HP}(z)$  are shown below



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# Simple IIR Digital Filters

- $H_{HP}(z)$  is a BR function for  $|\alpha| < 1$
- Example - Design a first-order highpass digital filter with a 3-dB cutoff frequency of  $0.8\pi$
- Now,  $\sin(\omega_c) = \sin(0.8\pi) = 0.587785$  and  $\cos(0.8\pi) = -0.80902$
- Therefore
$$\alpha = (1 - \sin \omega_c) / \cos \omega_c = -0.5095245$$

# Simple IIR Digital Filters

- Therefore,

$$\begin{aligned} H_{HP}(z) &= \frac{1 + \alpha}{2} \left( \frac{1 - z^{-1}}{1 - \alpha z^{-1}} \right) \\ &= 0.245238 \left( \frac{1 - z^{-1}}{1 + 0.5095245 z^{-1}} \right) \end{aligned}$$

# Simple IIR Digital Filters

## Bandpass IIR Digital Filters

can be also expressed  
as a LP+HP cascade

- A 2nd-order bandpass digital transfer function is given by

$$H_{BP}(z) = \frac{1-\alpha}{2} \left( \frac{1-z^{-2}}{1-\beta(1+\alpha)z^{-1}+\alpha z^{-2}} \right)$$

- Its squared magnitude function is

$$\begin{aligned} & |H_{BP}(e^{j\omega})|^2 \\ &= \frac{(1-\alpha)^2(1-\cos 2\omega)}{2[1+\beta^2(1+\alpha)^2+\alpha^2-2\beta(1+\alpha)^2\cos\omega+2\alpha\cos 2\omega]} \end{aligned}$$

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# Simple IIR Digital Filters

- $|H_{BP}(e^{j\omega})|^2$  goes to zero at  $\omega = 0$  and  $\omega = \pi$
- It assumes a maximum value of 1 at  $\omega = \omega_o$ , called the **center frequency** of the bandpass filter, where

$$\omega_o = \cos^{-1}(\beta)$$

- The frequencies  $\omega_{c1}$  and  $\omega_{c2}$  where  $|H_{BP}(e^{j\omega})|^2$  becomes 1/2 are called the **3-dB cutoff frequencies**

# Simple IIR Digital Filters

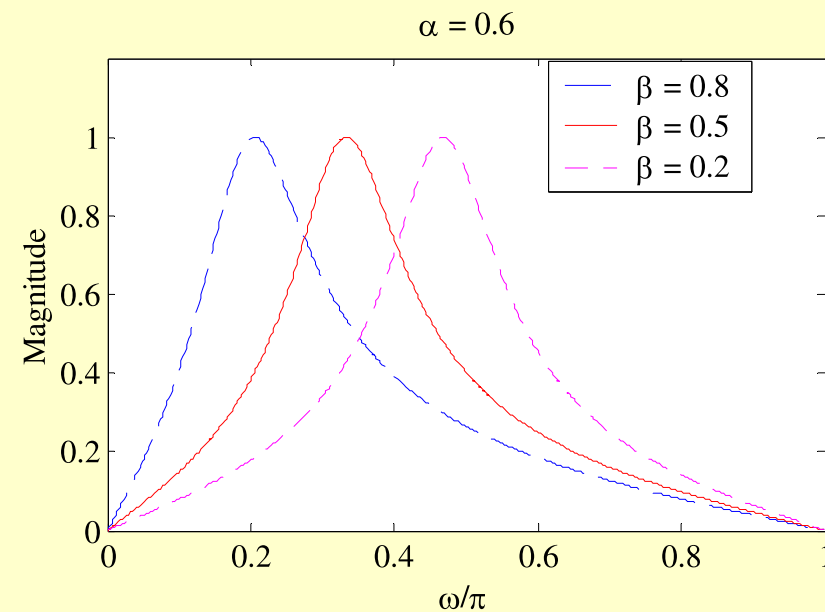
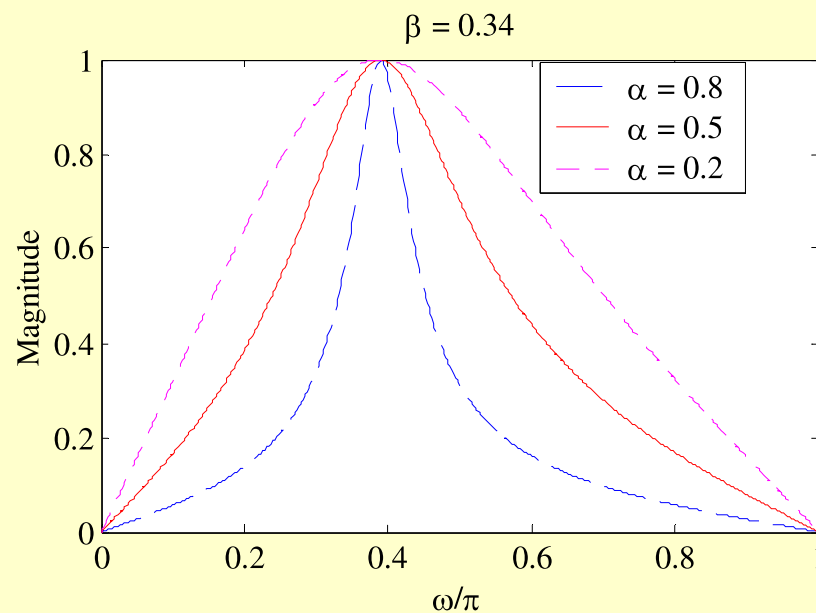
- The difference between the two cutoff frequencies, assuming  $\omega_{c2} > \omega_{c1}$  is called the **3-dB bandwidth** and is given by

$$B_w = \omega_{c2} - \omega_{c1} = \cos^{-1} \left( \frac{2\alpha}{1 + \alpha^2} \right)$$

- The transfer function  $H_{BP}(z)$  is a BR function if  $|\alpha| < 1$  and  $|\beta| < 1$

# Simple IIR Digital Filters

- Plots of  $|H_{BP}(e^{j\omega})|$  are shown below



# Simple IIR Digital Filters

- Example - Design a 2nd order bandpass digital filter with center frequency at  $0.4\pi$  and a 3-dB bandwidth of  $0.1\pi$

- Here  $\beta = \cos(\omega_o) = \cos(0.4\pi) = 0.309017$   
and

$$\frac{2\alpha}{1 + \alpha^2} = \cos(B_w) = \cos(0.1\pi) = 0.9510565$$

- The solution of the above equation yields:  
 $\alpha = 1.376382$  and  $\alpha = 0.72654253$

# Simple IIR Digital Filters

- The corresponding transfer functions are

$$H'_{BP}(z) = -0.18819 \frac{1 - z^{-2}}{1 - 0.7343424z^{-1} + 1.37638z^{-2}}$$

and

$$H''_{BP}(z) = 0.13673 \frac{1 - z^{-2}}{1 - 0.533531z^{-1} + 0.72654253z^{-2}}$$

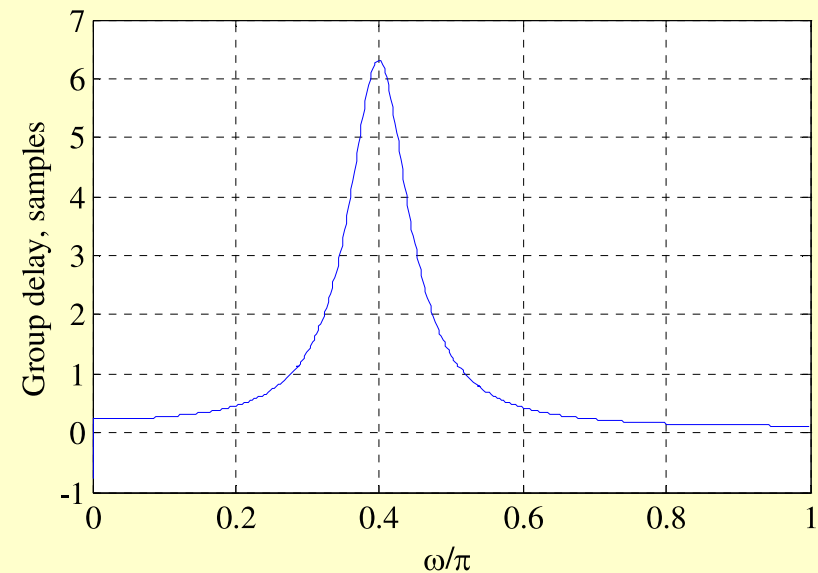
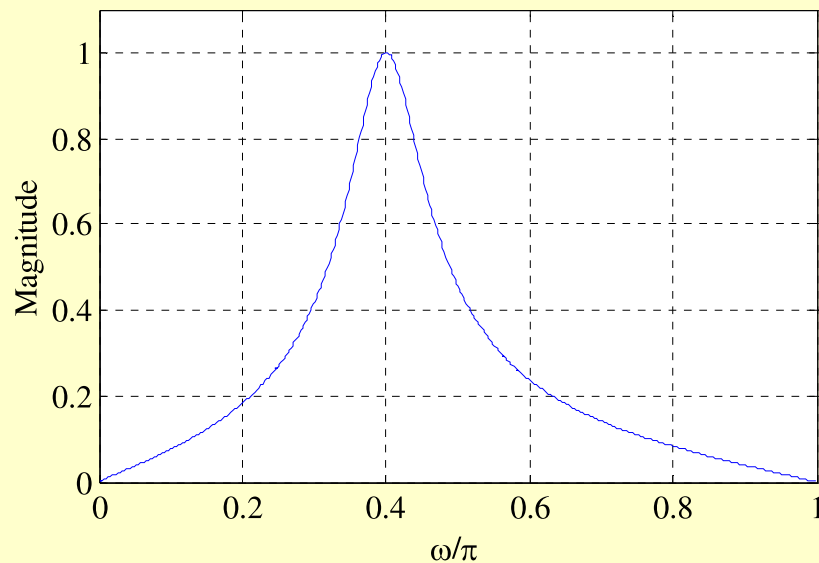
- The poles of  $H'_{BP}(z)$  are at  $z = 0.3671712 \pm j1.11425636$  and have a magnitude  $> 1$

# Simple IIR Digital Filters

- Thus, the poles of  $H'_{BP}(z)$  are outside the unit circle making the transfer function unstable
- On the other hand, the poles of  $H''_{BP}(z)$  are at  $z = 0.2667655 \pm j0.8095546$  and have a magnitude of 0.8523746
- Hence  $H''_{BP}(z)$  is BIBO stable
- Later we outline a simpler stability test

# Simple IIR Digital Filters

- Figures below show the plots of the magnitude function and the group delay of  $H''_{BP}(z)$



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# Simple IIR Digital Filters

## Bandstop IIR Digital Filters

- A 2nd-order bandstop digital filter has a transfer function given by

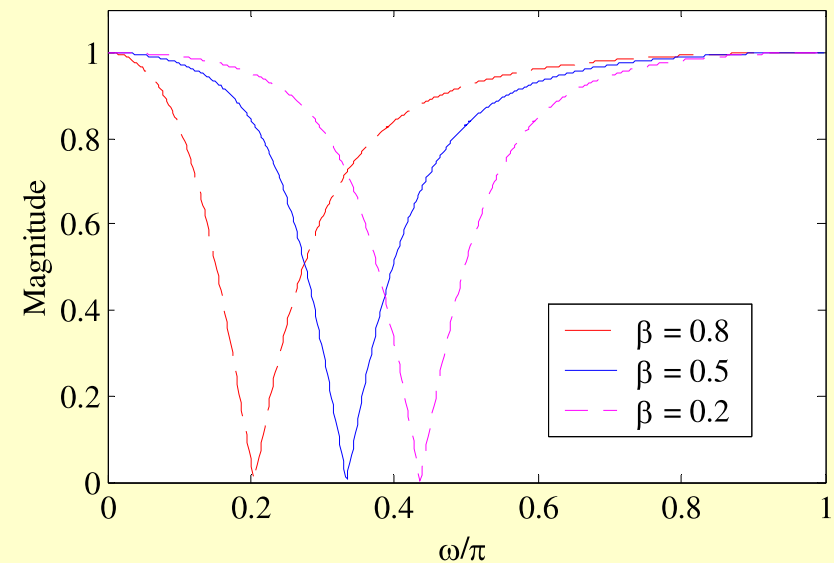
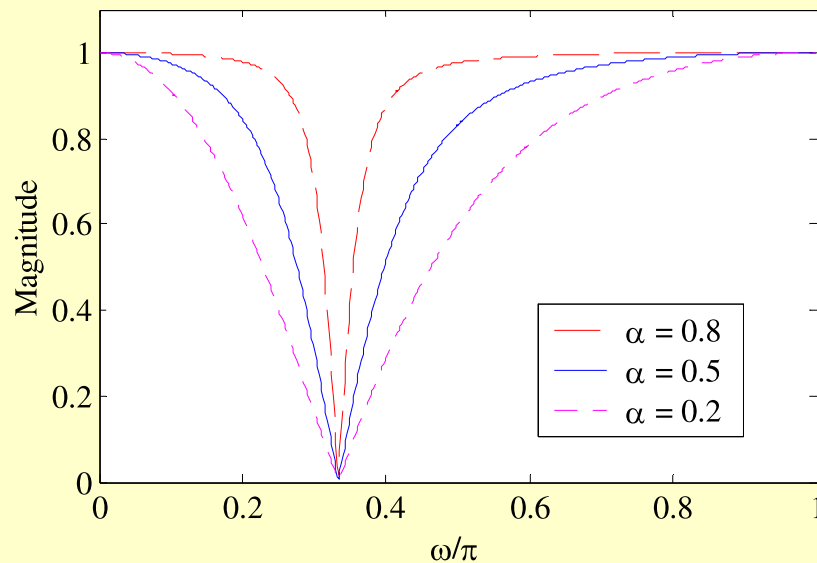
$$H_{BS}(z) = \frac{1 + \alpha}{2} \left( \frac{1 - 2\beta z^{-1} + z^{-2}}{1 - \beta(1 + \alpha)z^{-1} + \alpha z^{-2}} \right)$$

- The transfer function  $H_{BS}(z)$  is a BR function if  $|\alpha| < 1$  and  $|\beta| < 1$



# Simple IIR Digital Filters

- Its magnitude response is plotted below



# Simple IIR Digital Filters

- Here, the magnitude function takes the maximum value of 1 at  $\omega = 0$  and  $\omega = \pi$
- It goes to 0 at  $\omega = \omega_o$ , where  $\omega_o$ , called the **notch frequency**, is given by

$$\omega_o = \cos^{-1}(\beta)$$

- The digital transfer function  $H_{BS}(z)$  is more commonly called a **notch filter**

# Simple IIR Digital Filters

- The frequencies  $\omega_{c1}$  and  $\omega_{c2}$  where  $|H_{BS}(e^{j\omega})|^2$  becomes  $1/2$  are called the **3-dB cutoff frequencies**
- The difference between the two cutoff frequencies, assuming  $\omega_{c2} > \omega_{c1}$  is called the **3-dB notch bandwidth** and is given by

$$B_w = \omega_{c2} - \omega_{c1} = \cos^{-1}\left(\frac{2\alpha}{1 + \alpha^2}\right)$$

# Simple IIR Digital Filters

## Higher-Order IIR Digital Filters

- By cascading the simple digital filters discussed so far, we can implement digital filters with sharper magnitude responses
- Consider a cascade of  $K$  first-order lowpass sections characterized by the transfer function

$$H_{LP}(z) = \frac{1-\alpha}{2} \left( \frac{1+z^{-1}}{1-\alpha z^{-1}} \right)$$

# Simple IIR Digital Filters

- The overall structure has a transfer function given by

$$G_{LP}(z) = \left( \frac{1-\alpha}{2} \cdot \frac{1+z^{-1}}{1-\alpha z^{-1}} \right)^K$$

- The corresponding squared-magnitude function is given by

$$|G_{LP}(e^{j\omega})|^2 = \left[ \frac{(1-\alpha)^2 (1+\cos \omega)}{2(1+\alpha^2 - 2\alpha \cos \omega)} \right]^K$$

# Simple IIR Digital Filters

- To determine the relation between its 3-dB cutoff frequency  $\omega_c$  and the parameter  $\alpha$ , we set

$$\left[ \frac{(1-\alpha)^2(1+\cos\omega_c)}{2(1+\alpha^2-2\alpha\cos\omega_c)} \right]^K = \frac{1}{2}$$

which when solved for  $\alpha$ , yields for a stable  $G_{LP}(z)$ :

$$\alpha = \frac{1 + (1-C)\cos\omega_c - \sin\omega_c\sqrt{2C-C^2}}{1-C+\cos\omega_c}$$

# Simple IIR Digital Filters

where

$$C = 2^{(K-1)/K}$$

- It should be noted that the expression for  $\alpha$  given earlier reduces to

$$\alpha = \frac{1 - \sin \omega_c}{\cos \omega_c}$$

for  $K = 1$

# Simple IIR Digital Filters

- Example - Design a lowpass filter with a 3-dB cutoff frequency at  $\omega_c = 0.4\pi$  using a single first-order section and a cascade of 4 first-order sections, and compare their gain responses
- For the single first-order lowpass filter we have

$$\alpha = \frac{1 + \sin \omega_c}{\cos \omega_c} = \frac{1 + \sin(0.4\pi)}{\cos(0.4\pi)} = 0.1584$$



# Simple IIR Digital Filters

- For the cascade of 4 first-order sections, we substitute  $K = 4$  and get

$$C = 2^{(K-1)/K} = 2^{(4-1)/4} = 1.6818$$

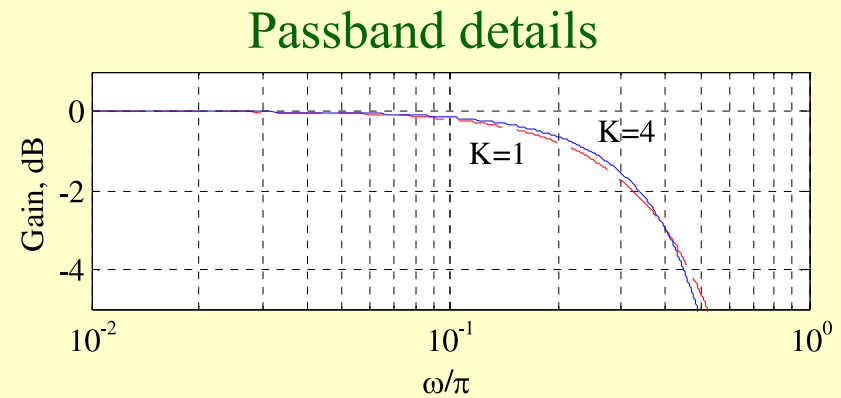
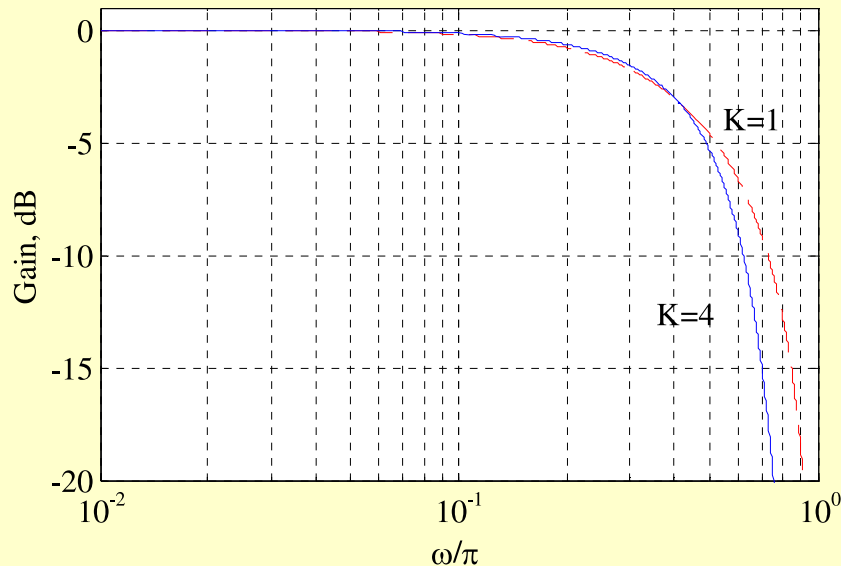
- Next we compute

$$\begin{aligned}\alpha &= \frac{1 + (1 - C)\cos\omega_c - \sin\omega_c\sqrt{2C - C^2}}{1 - C + \cos\omega_c} \\ &= \frac{1 + (1 - 1.6818)\cos(0.4\pi) - \sin(0.4\pi)\sqrt{2(1.6818) - (1.6818)^2}}{1 - 1.6818 + \cos(0.4\pi)} \\ &= -0.251\end{aligned}$$

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# Simple IIR Digital Filters

- The gain responses of the two filters are shown below
- As can be seen, cascading has resulted in a sharper roll-off in the gain response



# Digital Differentiators

- Employed to perform the differentiation operation on the discrete-time version of a continuous-time signal
- Frequency response of an ideal discrete-time differentiator is given by

$$H(e^{j\omega}) = j\omega \quad \text{for } 0 \leq |\omega| \leq \pi$$

which has a linear magnitude response from dc to  $\omega = \pi$

# Digital Differentiators

- A practical discrete-time differentiator is used to perform the differentiation operation in the low frequency range and is thus designed to have a linear magnitude response from dc to a frequency smaller than  $\pi$

# Simple FIR Digital Differentiators

**First-Difference** Differentiator is a first-order FIR discrete-time system with a time-domain input-output relation given by

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

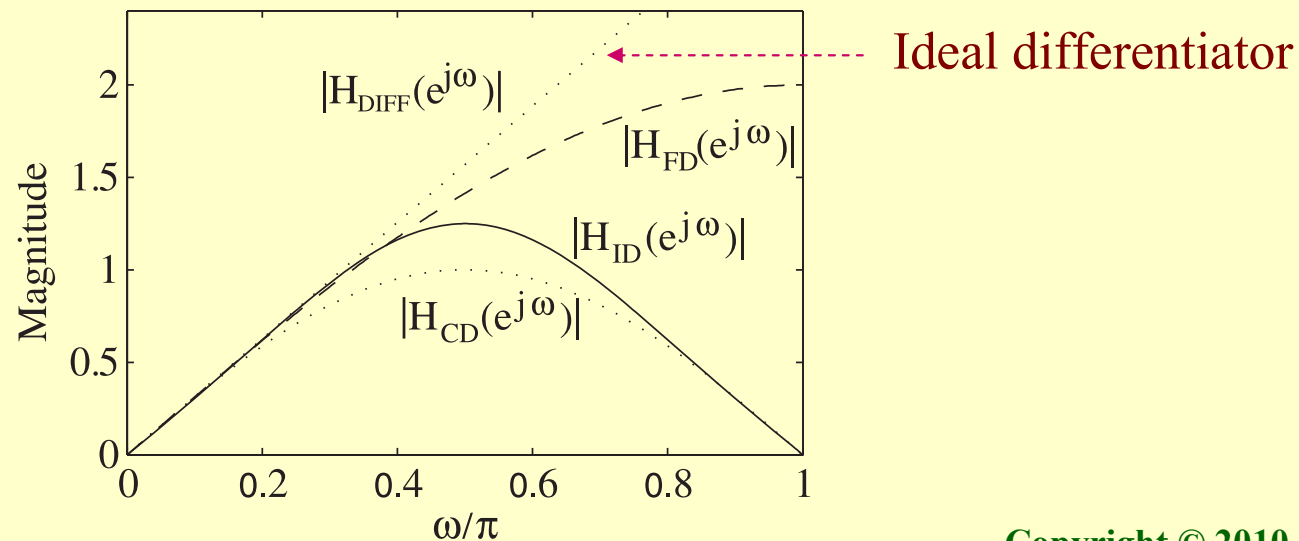
- Its transfer function is given by

$$H_{FD}(z) = 1 - z^{-1}$$

which is same as that of a first-order FIR highpass filter described earlier

# Simple FIR Digital Differentiators

- Main drawback of the first-difference differentiator is that it also amplifies the high frequency noise often present in many signals



# Simple FIR Digital Differentiators

**Central-Difference** Differentiator avoids the noise amplification problem of the first-difference differentiator

- Its time-domain input-output relation is

$$y[n] = \frac{1}{2}(x[n] - x[n-2])$$

- Its transfer function is given by

$$H_{CD}(z) = \frac{1}{2}(1 - z^{-2})$$

- It has a linear magnitude response in a very small low-frequency range

Can be derived also as the cascade of a basic lowpass and a basic highpass:  
 $h(n) = \text{conv}([1 \ 1], [1 \ -1])$

# Higher-Order FIR Digital Differentiator

- The time-domain input-output relation of a higher-order FIR digital differentiator is given by

$$y[n] = -\frac{1}{16}x[n] + x[n-2] - x[n-4] + \frac{1}{16}x[n-6]$$

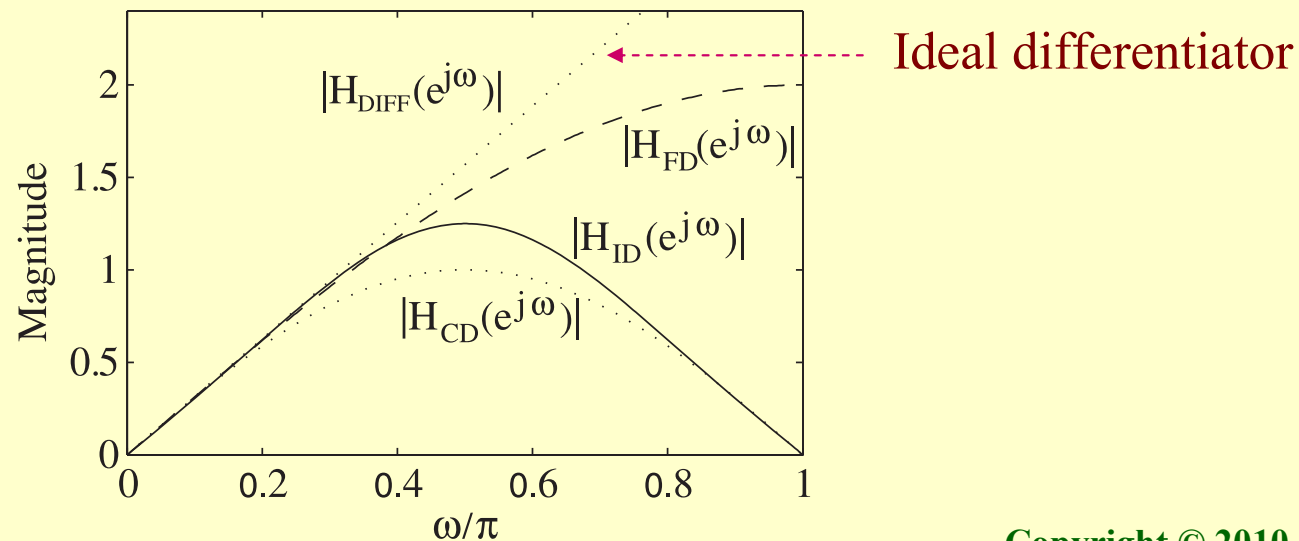
- Its transfer function is given by

$$H_{ID}(z) = -\frac{1}{16} + z^{-2} - z^{-4} + \frac{1}{16}z^{-6}$$



# Higher-Order FIR Digital Differentiator

- Its magnitude response, scaled by a factor 0.6 is shown below
- The frequency range of operation of this differentiator is from dc to  $\omega = 0.34\pi$



# Comb Filters

- The simple filters discussed so far are characterized either by a single passband and/or a single stopband
- There are applications where filters with multiple passbands and stopbands are required
- The **comb filter** is an example of such filters

# Comb Filters

- In its most general form, a comb filter has a frequency response that is a periodic function of  $\omega$  with a period  $2\pi/L$ , where  $L$  is a positive integer
- If  $H(z)$  is a filter with a single passband and/or a single stopband, a comb filter can be easily generated from it by replacing each delay in its realization with  $L$  delays resulting in a structure with a transfer function given by  $G(z) = H(z^L)$

For  $\omega$  in  $[0, 2\pi]$  we move  $L$  times along the unit circle

# Comb Filters

- If  $|H(e^{j\omega})|$  exhibits a peak at  $\omega_p$ , then  $|G(e^{j\omega})|$  will exhibit  $L$  peaks at  $\omega_p k/L$ ,  $0 \leq k \leq L-1$  in the frequency range  $0 \leq \omega < 2\pi$
- Likewise, if  $|H(e^{j\omega})|$  has a notch at  $\omega_o$ , then  $|G(e^{j\omega})|$  will have  $L$  notches at  $\omega_o k/L$ ,  $0 \leq k \leq L-1$  in the frequency range  $0 \leq \omega < 2\pi$
- A comb filter can be generated from either an FIR or an IIR prototype filter

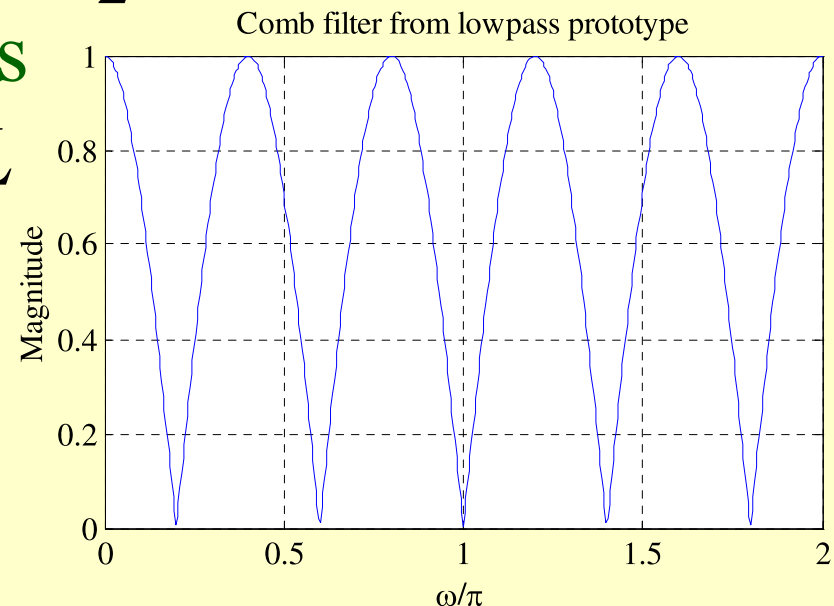
# FIR Comb Filters

- For example, the comb filter generated from the prototype lowpass FIR filter  $H_0(z) = \frac{1}{2}(1 + z^{-1})$  has a transfer function

$$G_0(z) = H_0(z^L) = \frac{1}{2}(1 + z^{-L})$$

Impulse response?

- $|G_0(e^{j\omega})|$  has  $L$  notches at  $\omega = (2k+1)\pi/L$  and  $L$  peaks at  $\omega = 2\pi k/L$ ,  $0 \leq k \leq L-1$ , in the frequency range  $0 \leq \omega < 2\pi$



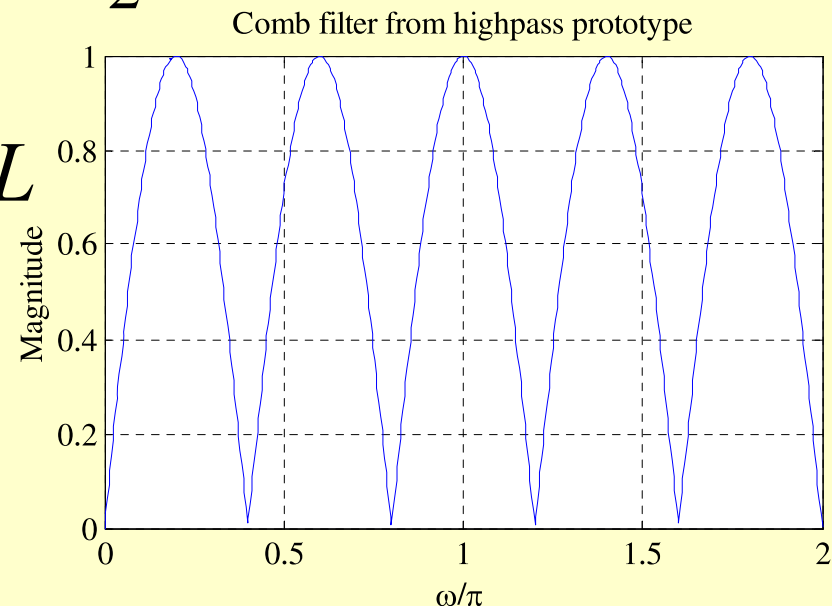
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# FIR Comb Filters

- For example, the comb filter generated from the prototype highpass FIR filter  $H_1(z) = \frac{1}{2}(1 - z^{-1})$  has a transfer function

$$G_1(z) = H_1(z^L) = \frac{1}{2}(1 - z^{-L})$$

- $|G_1(e^{j\omega})|$  has  $L$  peaks at  $\omega = (2k+1)\pi/L$  and  $L$  notches at  $\omega = 2\pi k/L$ ,  $0 \leq k \leq L-1$ , in the frequency range  $0 \leq \omega < 2\pi$



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# FIR Comb Filters

- Depending on applications, comb filters with other types of periodic magnitude responses can be easily generated by appropriately choosing the prototype filter
- For example, the  $M$ -point moving average filter

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

has been used as a prototype

# FIR Comb Filters

- This filter has a peak magnitude at  $\omega = 0$ , and  $M - 1$  notches at  $\omega = 2\pi\ell / M, 1 \leq \ell \leq M - 1$
- The corresponding comb filter has a transfer function

$$G(z) = \frac{1 - z^{-LM}}{M(1 - z^{-L})}$$

whose magnitude has  $L$  peaks at  $\omega = 2\pi k/L, 0 \leq k \leq L - 1$  and  $L(M - 1)$  notches at  $\omega = 2\pi k/LM, 1 \leq k \leq L(M - 1)$



## IIR Comb Filters

- The transfer functions of the simplest forms of the prototype IIR filter are given by

$$H_0(z) = K \frac{1 - z^{-1}}{1 - \alpha z^{-1}}, \quad H_1(z) = K \frac{1 + z^{-1}}{1 - \alpha z^{-1}}$$

where  $|\alpha| < 1$  for stability

- **Note:**  $H_0(z)$  is a highpass filter with a zero at  $z = 1$  and  $H_1(z)$  is a lowpass filter with a zero at  $z = -1$

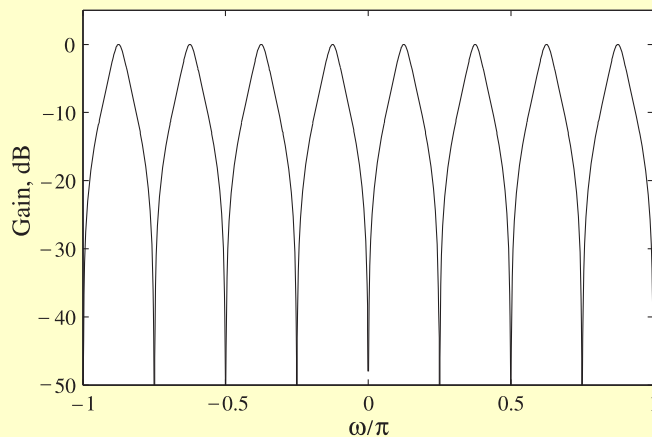
# IIR Comb Filters

- For a maximum gain of 0 dB, the scale factor  $K$  of  $H_0(z)$  should be set equal to  $(1 + \alpha) / 2$  and the scale factor  $K$  of  $H_1(z)$  should be set equal to  $(1 - \alpha) / 2$
- The corresponding transfer functions of the comb filters of order  $L$  are

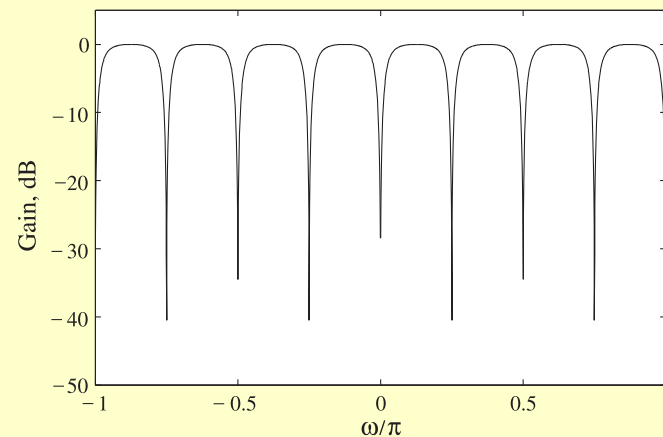
$$G_0(z) = K \frac{1 - z^{-L}}{1 - \alpha z^{-L}}, \quad G_1(z) = K \frac{1 + z^{-L}}{1 - \alpha z^{-L}}$$

# IIR Comb Filters

- Gain responses of the IIR comb filters generated from  $H_0(z)$  and  $H_1(z)$  for  $L = 8$  are shown below



$$G_0(z) = H_0(z^8)$$



$$G_1(z) = H_1(z^8)$$

# Complementary Transfer Functions

- A set of digital transfer functions with complementary characteristics often finds useful applications in practice
- useful complementary relations are described next along with some applications

# Allpass Complementary Transfer Functions

- A set of  $M$  digital transfer functions,  $\{H_i(z)\}$ ,  $0 \leq i \leq M-1$ , is defined to be **allpass-complementary** of each other, if the sum of their transfer functions is equal to an allpass function, i.e.,

$$\sum_{i=0}^{M-1} H_i(z) = A(z)$$

# Power-Complementary Transfer Functions

- A set of  $M$  digital transfer functions,  $\{H_i(z)\}$ ,  $0 \leq i \leq M - 1$ , is defined to be **power-complementary** of each other, if the sum of their square-magnitude responses is equal to a constant  $K$  for all values of  $\omega$ , i.e.,

$$\sum_{i=0}^{M-1} |H_i(e^{j\omega})|^2 = K, \quad \text{for all } \omega$$

# Power-Complementary Transfer Functions

- For a pair of power-complementary transfer functions,  $H_0(z)$  and  $H_1(z)$ , the frequency  $\omega_o$  where  $|H_0(e^{j\omega_o})|^2 = |H_1(e^{j\omega_o})|^2 = 0.5$ , is called the **cross-over frequency**
- At this frequency the gain responses of both filters are 3-dB below their maximum values
- As a result,  $\omega_o$  is called the **3-dB cross-over frequency**

# Power-Complementary Transfer Functions

- Example - Consider the two transfer functions  $H_0(z)$  and  $H_1(z)$  given by

$$H_0(z) = \frac{1}{2}[\mathcal{A}_0(z) + \mathcal{A}_1(z)]$$

$$H_1(z) = \frac{1}{2}[\mathcal{A}_0(z) - \mathcal{A}_1(z)]$$

where  $\mathcal{A}_0(z)$  and  $\mathcal{A}_1(z)$  are stable allpass transfer functions

- Note that  $H_0(z) + H_1(z) = \mathcal{A}_0(z)$
- Hence,  $H_0(z)$  and  $H_1(z)$  are allpass complementary

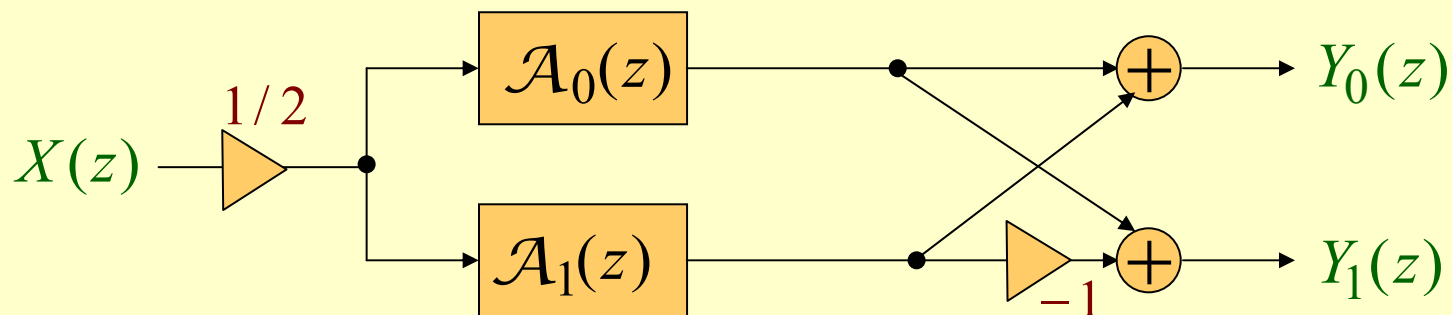


# Power-Complementary Transfer Functions

- It can be shown that  $H_0(z)$  and  $H_1(z)$  are also power-complementary
- Moreover,  $H_0(z)$  and  $H_1(z)$  are bounded-real transfer functions

# Doubly-Complementary Transfer Functions

- A pair of doubly-complementary IIR transfer functions,  $H_0(z)$  and  $H_1(z)$ , with a sum of allpass decomposition can be simply realized as indicated below



$$H_0(z) = \frac{Y_0(z)}{X(z)}$$

$$H_1(z) = \frac{Y_1(z)}{X(z)}$$

# Doubly-Complementary Transfer Functions

- Example - The first-order lowpass transfer function

$$H_{LP}(z) = \frac{1-\alpha}{2} \left( \frac{1+z^{-1}}{1-\alpha z^{-1}} \right)$$

can be expressed as

$$H_{LP}(z) = \frac{1}{2} \left( 1 + \frac{-\alpha + z^{-1}}{1-\alpha z^{-1}} \right) = \frac{1}{2} [\mathcal{A}_0(z) + \mathcal{A}_1(z)]$$

where

$$\mathcal{A}_0(z) = 1, \quad \mathcal{A}_1(z) = \frac{-\alpha + z^{-1}}{1-\alpha z^{-1}}$$

# Doubly-Complementary Transfer Functions

- Its power-complementary highpass transfer function is thus given by

$$\begin{aligned} H_{HP}(z) &= \frac{1}{2}[\mathcal{A}_0(z) - \mathcal{A}_1(z)] = \frac{1}{2}\left(1 - \frac{-\alpha + z^{-1}}{1 - \alpha z^{-1}}\right) \\ &= \frac{1 + \alpha}{2} \left( \frac{1 - z^{-1}}{1 - \alpha z^{-1}} \right) \end{aligned}$$

- The above expression is precisely the first-order highpass transfer function described earlier

# Doubly-Complementary Transfer Functions

- Figure below demonstrates the allpass complementary property and the power complementary property of  $H_{LP}(z)$  and  $H_{HP}(z)$

