

9.13 Let  $H(z)$  be the T.F. of a lowpass digital filter with a passband edge at  $\omega_p$ , stopband edge at  $\omega_s$ , passband ripple of  $\delta_p$ , and stopband ripple of  $\delta_s$ , as indicated in Fig 9.1. Consider a cascade of two identical filters with T.F.  $H(z)$ . What are the passband/stopband ripples of the cascade at  $\omega_p$  and  $\omega_s$ , respectively? Generalize results to  $M$  identical sections.

Let  $G(z) = H^2(z)$  and  $G(e^{j\omega}) = H^2(e^{j\omega})$   
 $|G(e^{j\omega})| = |H^2(e^{j\omega})| = |H(e^{j\omega})|^2$

the passband/stopband ripples of  $G(e^{j\omega})$  will be  $\delta_{p,2} = 1 - (1 - \delta_p)^2$  and  $\delta_{s,2} = (\delta_s)^2$ .

For  $M$  sections,  $\delta_{p,M} = 1 - (1 - \delta_p)^M$   $\delta_{s,M} = (\delta_s)^M$

9.17 Let  $H_a(s)$  be a real-coefficient causal and stable analog T.F. with a magnitude response bounded above by 1. Show that the digital T.F.  $G(z)$  obtained by bilinear transformation of  $H_a(s)$  is a bounded real function.

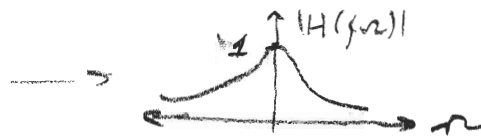
$H_a(s)$  is causal and stable ...

$$G(z) = H_a(s) \Big|_{s=j\frac{2}{T}\tan(\frac{\omega}{2})} = H_a(j\frac{2}{T}\tan(\frac{\omega}{2}))$$

$\max(|H_a(j\omega)|) = 1$

$$|G(e^{j\omega})| = |H_a(j\frac{2}{T}\tan(\frac{\omega}{2}))|$$

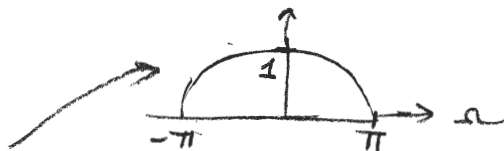
Example:  $H_a(s) = \frac{\omega_c}{s + \omega_c}$



$$G(e^{j\omega}) = \frac{\omega_c}{j\frac{2}{T}\tan(\frac{\omega}{2}) + \omega_c}$$

$$|G(e^{j\omega})| = \frac{\omega_c}{\sqrt{\frac{4}{T^2}\tan^2(\frac{\omega}{2}) + \omega_c^2}}$$

$\max(|G(e^{j\omega})|) = 1$  @  $(\omega = 0)$



9.21 Let  $H_{LP}(z)$  be an IIR lowpass T.F with a zero/pole at  $z = z_k$ . Let  $H_B(\hat{z})$  denote the bandpass T.F obtained by applying the low-to-bandpass transformation, which moves the zero (pole) at  $z = z_k$  of  $H_{LP}(z)$  to a new location at  $\hat{z} = \hat{z}_k$ . Express  $\hat{z}_k$  in terms of  $z_k$  if  $H_{LP}(z)$  has a zero at  $z = -1$ , show that  $H_B(\hat{z})$  also has a zero at  $z = \pm 1$ .

Let us use the low-to-bandpass transformation (Table 9.1)

$$\hat{z}^{-1} = -\frac{\hat{z}_k^{-2} - a_1 \hat{z}_k^{-1} + a_2}{a_2 \hat{z}_k^{-2} - a_1 \hat{z}_k^{-1} + 1}$$

$$a_1 = \frac{z+p}{p+1}$$

$$a_2 = \frac{p-1}{p+1}$$

$$z = -\frac{a_2 \hat{z}_k^2 - a_1 \hat{z}_k + 1}{\hat{z}_k^2 - a_1 \hat{z}_k + a_2} = -\frac{a_2 - a_1 \hat{z}_k + \hat{z}_k^2}{1 - a_1 \hat{z}_k + a_2 \hat{z}_k^2}$$

A zero/pole located at  $(z - z_k)$  will be mapped to:

$$-\frac{a_2 - a_1 \hat{z}_k + \hat{z}_k^2}{1 - a_1 \hat{z}_k + a_2 \hat{z}_k^2} - z_k = 0 \quad -a_2 + a_1 \hat{z}_k - \hat{z}_k^2 - z_k + a_1 z_k \hat{z}_k - a_2 z_k \hat{z}_k^2 = 0$$

$$(-1 - a_2 z_k) \hat{z}_k^2 + a_1 (1 + z_k) \hat{z}_k + (-a_2 - z_k) = 0$$

$$\hat{z}_k^2 - \frac{a_1 (1 + z_k)}{1 + a_2 z_k} \hat{z}_k + \frac{(a_2 + z_k)}{(1 + a_2 z_k)} = 0$$

$$\hat{z}_k = \frac{a_1 (1 + z_k)}{(1 + a_2 z_k)} \pm \sqrt{\left[ \frac{a_1 (1 + z_k)}{(1 + a_2 z_k)} \right]^2 - \frac{4(a_2 + z_k)}{(1 + a_2 z_k)}}$$

$$= \frac{a_1 (1 + z_k)}{2(1 + a_2 z_k)} \pm \sqrt{\left[ \frac{a_1 (1 + z_k)}{2(1 + a_2 z_k)} \right]^2 - \frac{(a_2 + z_k)}{1 + a_2 z_k}}$$

Let us find out where the zero/pole at  $z = -1$  maps to:

$$\hat{z}_k = \pm \sqrt{\frac{1 - a_2}{1 - a_2}} = \pm 1$$

9.26 Design a bandpass filter with center frequency at  $\hat{\omega}_0 = 0.5\pi$  by transforming the bandpass in Eq. 7.79 using lowpass-to-lowpass.

$$H_{BP}(z) = 0.136728736 \left( \frac{1 - z^{-2}}{1 - 0.53353098z^{-1} + 0.726542528z^{-2}} \right)$$

Plotting  $H_{BP}(z)$  we know that  $\omega_0 = 0.4\pi$   
 So, using low-to-low transformation (Table 9.1)

$$\lambda = \frac{\sin(-0.05\pi)}{\sin(0.45\pi)} = -0.1584$$

$$G_{BP}(z) = H_{BP}(z) \Big|_{z^{-1} = \frac{\hat{z}^{-1} - \lambda}{1 - \lambda\hat{z}^{-1}}} = \frac{a \left[ 1 - \frac{(\hat{z}^{-1} - \lambda)^2}{(1 - \lambda\hat{z}^{-1})^2} \right]}{1 - b \frac{(\hat{z}^{-1} - \lambda)}{(1 - \lambda\hat{z}^{-1})} + c \frac{(\hat{z}^{-1} - \lambda)^2}{(1 - \lambda\hat{z}^{-1})^2}}$$

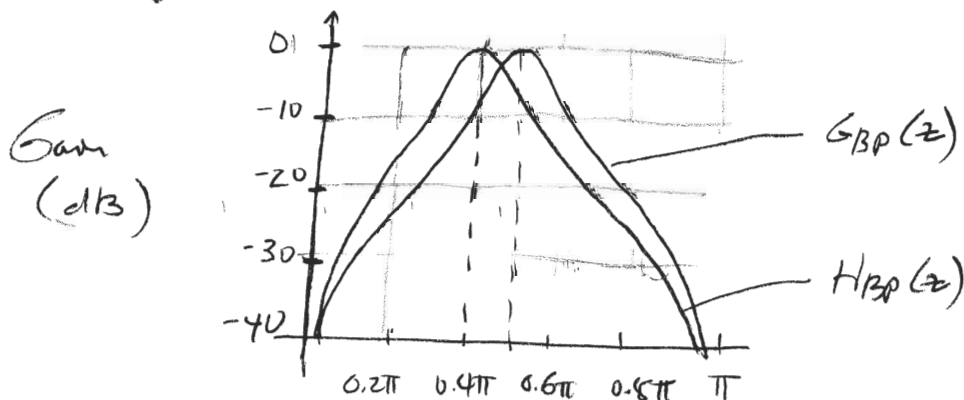
$$= \frac{a[(1 - \lambda\hat{z}^{-1})^2 - (\hat{z}^{-1} - \lambda)^2]}{(1 - \lambda\hat{z}^{-1})^2 - b(\hat{z}^{-1} - \lambda)(1 - \lambda\hat{z}^{-1}) + c(\hat{z}^{-1} - \lambda)^2}$$

$$= \frac{a[(1 - 2\lambda\hat{z}^{-1} + \lambda^2\hat{z}^{-2}) - (\hat{z}^{-2} - 2\lambda\hat{z}^{-1} + \lambda^2)]}{1 - 2\lambda\hat{z}^{-1} + \lambda^2\hat{z}^{-2} - b(\hat{z}^{-1} - \lambda - \lambda\hat{z}^{-2} + \lambda^2\hat{z}^{-1}) + c(\hat{z}^{-2} - 2\lambda\hat{z}^{-1} + \lambda^2)}$$

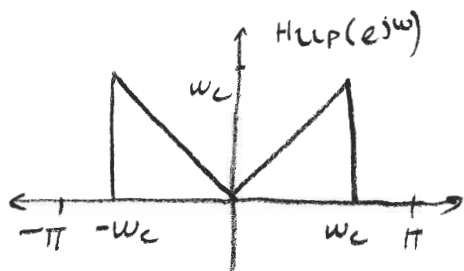
$$= \frac{a[(1 + \lambda^2) + (\lambda^2 - 1)\hat{z}^{-2}]}{(1 + b\lambda + c\lambda^2) + (-2\lambda - b - b\lambda^2 - 2c\lambda)\hat{z}^{-1} + (\lambda^2 + b\lambda + c)\hat{z}^{-2}}$$

$$= \frac{0.1333 - 0.1333\hat{z}^{-2}}{0.9337 + 0.6672\hat{z}^{-1} + 0.6672\hat{z}^{-2}}$$

Using MATLAB to plot gains of both filters.



10.10 Determine the impulse response  $h_{LP}[n]$  of a zero-phase ideal linear passband lowpass filter characterized by a frequency response.



$$H_{LP}(e^{j\omega}) = \begin{cases} |\omega| & |\omega| \leq \omega_c \\ 0 & \text{o.w.} \end{cases}$$

$$\begin{aligned} h_{LP}[n] &= \frac{1}{2\pi} \left[ \int_{-\omega_c}^0 \omega e^{j\omega n} d\omega + \int_0^{\omega_c} \omega e^{j\omega n} d\omega \right] = \frac{1}{2\pi} \left[ \left( \frac{\omega e^{j\omega n}}{jn} + \frac{e^{j\omega n}}{n^2} \right) \Big|_{-\omega_c}^0 + \right. \\ &\quad \left. \left( \frac{\omega e^{j\omega n}}{jn} + \frac{e^{j\omega n}}{n^2} \right) \Big|_0^{\omega_c} \right] = \frac{1}{2\pi} \left[ -\frac{1}{n^2} - \frac{\omega_c e^{-j\omega_c n}}{jn} + \frac{e^{-j\omega_c n}}{n^2} + \right. \\ &\quad \left. \frac{\omega_c e^{j\omega_c n}}{jn} + \frac{e^{j\omega_c n}}{n^2} - \frac{1}{n^2} \right] = \frac{1}{2\pi} \left[ \frac{\omega_c e^{j\omega_c n} - \omega_c e^{-j\omega_c n}}{jn} + \frac{e^{j\omega_c n} + e^{-j\omega_c n}}{n^2} - \frac{2}{n^2} \right] \\ &= \frac{\omega_c}{\pi n} \sin(\omega_c n) + \frac{\cos(\omega_c n)}{\pi n^2} - \frac{1}{\pi n} \end{aligned}$$

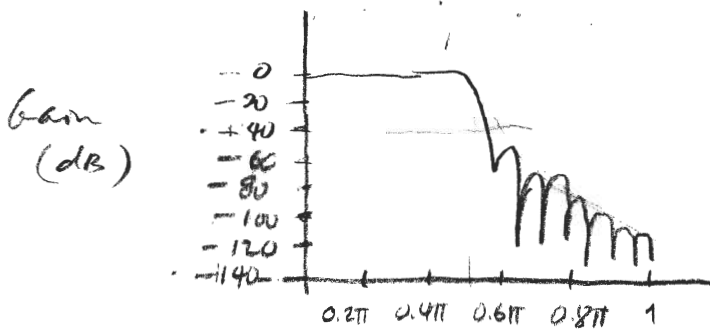
10.15 For each of the lowpass filter specifications given below, design an FIR filter with the smallest length meeting the specifications using the window-based approach, and plot its magnitude response using MATLAB:

a)  $\omega_p = 0.47\pi$ ,  $\omega_s = 0.59\pi$ ,  $\delta_p = 0.001$ ,  $\delta_s = 0.007$   
 $\alpha_s = -20 \log_{10}(0.007) = 43.098$  dB

From Table 10.2, we can achieve minimum stopband attenuation using Hann, Hamming or Blackman window.

Hann will have the lowest filter length

$$0.12\pi = \frac{3.11\pi}{M} \quad M = 25.917 \quad N = \lceil 2M + 1 \rceil = 53$$

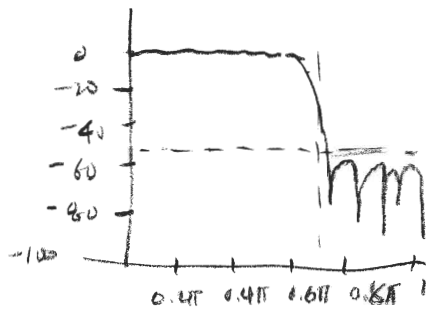


(b)  $\omega_p = 0.61\pi$ ,  $\omega_s = 0.78\pi$ ,  $\delta_p = 0.001$ ,  $\delta_s = 0.002$ , 1

$\alpha_s = -20 \log_{10}(0.002) = 53.979$  dB

From Table 10.2 we see that Hamming or Blackman can provide this attenuation.

$M = \frac{3.32\pi}{0.17\pi} = 19.529$       $N_{\text{Ham}} = \lceil 2M+1 \rceil = 41$



$L=3, M=1, N=1$   
 $10.29 \quad L=4, M=2, N=1$

w/e

In problem 9.3, we considered filtering by a cascade of a number of identical filters. In the case of an FIR filter  $H(z)$  with a symmetric impulse response, improved passband/stopband performances can be achieved by employing the FILTER SHARPENING APPROACH in which  $G(z)$  is implemented as

$$G(z) = \sum_{l=1}^L \alpha_l [H(z)]^l \quad \{\alpha_l\} \text{ are real coeffs}$$

It follows that  $G(z)$  is also an FIR filter with a symmetric impulse response. Let  $x$  denote a specific amplitude response of  $H(z)$  at freq.  $\omega$ . If we denote the value of  $G(z)$  at this value of  $\omega$  as  $P(x)$ , then

$$P(x) = \sum_{l=1}^L \alpha_l x^l$$

For BR T.F.  $H(z)$ ,  $0 \leq x \leq 1$ , where  $x=0$  is in stopband and  $x=1$  is in passband. If we desire  $P(0)=0$  &  $P(1)=1$  and constraining the slope at  $x=0$  &  $x=1$ .

To improve the performance of  $G(z)$  in stop bands and pass bands.

$$\left. \begin{aligned} \frac{d^k P(x)}{dx^k} \Big|_{x=0} = 0, \quad k=1, 2, \dots, n \\ \frac{d^k P(x)}{dx^k} \Big|_{x=1} = 0, \quad k=1, 2, \dots, m \end{aligned} \right\} \text{where } m+n=L-1$$

a) Determine  $\{\alpha_i\}$  for  $L=3$

$$L=3 = m+n+1 \text{ then pick } m=n=1$$

$$P(x) = \alpha_1 x + \alpha_2 x^2 + \alpha_3 x^3$$

$$\checkmark P(0) = 0 \text{ satisfied}$$

$$\checkmark P(1) = \alpha_1 + \alpha_2 + \alpha_3 = 1$$

$$\checkmark \frac{dP(x)}{dx} \Big|_{x=0} = \alpha_1 + 2\alpha_2 x + 3\alpha_3 x^2 \Big|_{x=0} = \alpha_1 = 0$$

$$\checkmark \frac{dP(x)}{dx} \Big|_{x=1} = \alpha_1 + 2\alpha_2 x + 3\alpha_3 x^2 \Big|_{x=1} = \alpha_1 + 2\alpha_2 + 3\alpha_3 = 0$$

Solving these eqns

$$\alpha_1 = 0 \quad \alpha_2 = 1 - \alpha_3 = \underline{3}$$

$$\alpha_2 + \alpha_3 = 1 \quad 2 - 2\alpha_3 + 3\alpha_3 = 0$$

$$2\alpha_2 + 3\alpha_3 = 0 \quad \alpha_3 = \underline{-2}$$

$$\underline{P(x) = 3x^2 - 2x^3}$$

b) Determine  $\{\alpha_i\}$  for  $L=4$

$$L=4 = m+n+1 \text{ choose } m=2 \ \& \ n=1$$

$$P(x) = \alpha_1 x + \alpha_2 x^2 + \alpha_3 x^3 + \alpha_4 x^4$$

$$\checkmark P(0) = 0$$

$$\checkmark P(1) = \alpha_1 + \alpha_2 + \alpha_3 + \alpha_4 = 1$$

$$\checkmark \frac{dP(x)}{dx} \Big|_{x=0} = \alpha_1 + 2\alpha_2 x + 3\alpha_3 x^2 + 4\alpha_4 x^3 \Big|_{x=0} = 0 \Rightarrow \underline{\alpha_1 = 0}$$

$$\checkmark \frac{d^2 P(x)}{dx^2} \Big|_{x=0} = 2\alpha_2 + 6\alpha_3 x + 12\alpha_4 x^2 \Big|_{x=0} = 0 \Rightarrow 2\alpha_2 = 0 \quad \underline{\alpha_2 = 0}$$

$$\checkmark \frac{dP(x)}{dx} \Big|_{x=1} = 0 \Rightarrow \alpha_1 + 2\alpha_2 + 3\alpha_3 + 4\alpha_4 = 0$$

Determine  $\{\alpha_k\}$  for  $l=4$

$$L = 4 = m+1+1 \quad m=1 \quad \& \quad n=2$$

$$P(x) = \alpha_1 x + \alpha_2 x^2 + \alpha_3 x^3 + \alpha_4 x^4$$

$$P(0) = 0$$

$$P(1) = \alpha_1 + \alpha_2 + \alpha_3 + \alpha_4 = 1$$

$$\frac{dP(x)}{dx} \Big|_{x=0} = \alpha_1 + 2\alpha_2 x + 3\alpha_3 x^2 + 4\alpha_4 x^3 \Big|_{x=0} \Rightarrow \alpha_1 = 0$$

$$\frac{dP(x)}{dx} \Big|_{x=1} = \alpha_1 + 2\alpha_2 + 3\alpha_3 + 4\alpha_4 = 0$$

$$\frac{d^2 P(x)}{dx^2} \Big|_{x=1} = 2\alpha_2 + 6\alpha_3 x + 12\alpha_4 x^2 \Big|_{x=1} = 0$$

$$2\alpha_2 + 6\alpha_3 + 12\alpha_4 = 0$$

$$\alpha_2 + \alpha_3 + \alpha_4 = 1$$

$$\alpha_1 = 0$$

$$2\alpha_2 + 3\alpha_3 + 4\alpha_4 = 0$$

$$\alpha_2 = 6$$

$$2\alpha_2 + 6\alpha_3 + 12\alpha_4 = 0$$

$$\alpha_3 = -8$$

$$\alpha_4 = 3$$

$$\underline{P(x) = 6x^2 - 8x^3 + 3x^4}$$

$$\begin{aligned} \alpha_3 + \alpha_4 &= 1 \\ 3\alpha_3 + 4\alpha_4 &= 0 \end{aligned}$$

$$\alpha_3 = 1 - \alpha_4 = 4$$

$$3 - 3\alpha_4 + 4\alpha_4 = 0$$

$$\alpha_4 = -3$$

$$P(x) = 4x^3 - 3x^4$$

**M10.22** The `fvfir2` function is employed to design FIR filters with arbitrarily shaped magnitude responses. Using this function, design a FIR of order 70 with 3 different magnitude levels:

$$0.2 \rightarrow 0 - 0.35\pi$$

$$1.0 \rightarrow 0.4\pi - 0.7\pi$$

$$0.6 \rightarrow 0.72\pi - \pi$$

Plot gain response of filter.

$$Fr = [0 \ 0.35 \ 0.4 \ 0.7 \ 0.72 \ 1];$$

$$Mag = [0.2 \ 0.2 \ 1 \ 1 \ 0.6 \ 0.6];$$

$$N = 70$$

$$b = \text{fvfir2}(N, Fr, Mag);$$

$$[H, w] = \text{freqz}(b, 1, 512);$$

$$\text{plot}(w/\pi, \text{abs}(H)); \text{xlabel}('w'); \text{ylabel}('Mag');$$

$$\text{axis}([0 \ 1 \ 0 \ 1.2]);$$

