

# EE538

## Module 15

DSPI

### Outline :

- Bilinear Transform Examples  
from old Exam 3's

- Design of IIR Filters via Bilinear Transform
  - Example 1.
  - 2<sup>nd</sup> order Butterworth LPF with a 3-dB cut-off at  $\Omega_c = 1$  has the transfer function
- $$H_a(s) = \frac{1}{s^2 + \sqrt{2}s + 1}$$

Use bilinear transform

$$S = c \frac{z-1}{z+1}$$

to design a digital filter  
with 3 dB cut-off at

$$\omega_c = \pi/2 \text{ rads./sec.}$$

(a)  $\Omega_c = c \tan(\omega_c/2)$

$$1 = c \tan(\frac{\pi}{2}/2)$$

• since  $\tan(\pi/4) = 1 \Rightarrow c=1$

$$\begin{aligned} H(z) &= H_a(s) \Big|_{s=\frac{z-1}{z+1}} \\ &= \frac{1}{\frac{(z-1)^2}{(z+1)^2} + \sqrt{2} \frac{z-1}{z+1} + 1} \cdot \frac{(z+1)^2}{(z+1)^2} \\ &= \frac{(z+1)^2}{(z-1)^2 + \sqrt{2}(z^2-1) + (z+1)^2} \end{aligned}$$

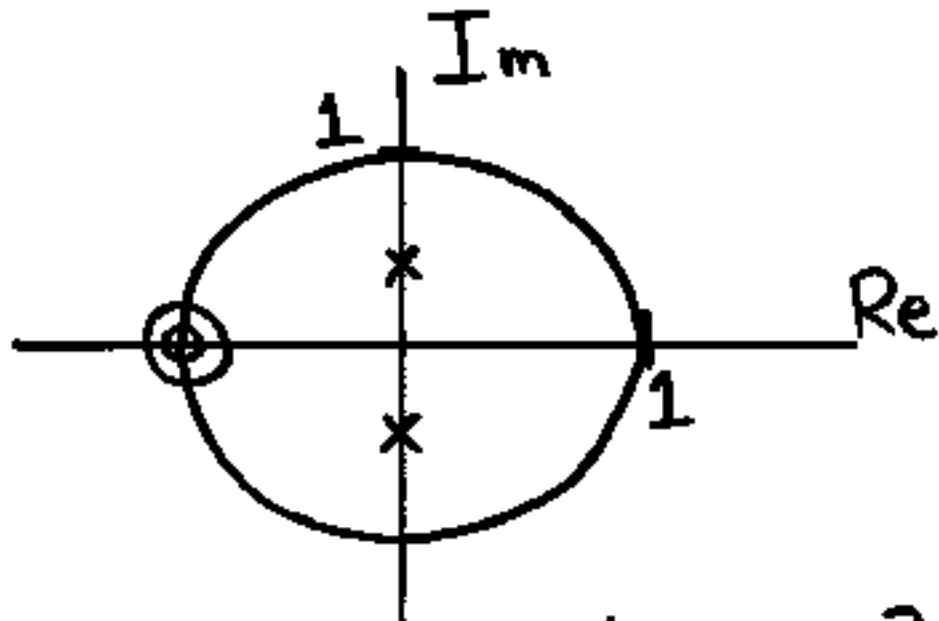
$$H(z) = \frac{(z+1)^2}{z^2 - 2z + 1 + \sqrt{2}z^2 - \sqrt{2}}$$

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

• poles:

$$P_k = \pm j \sqrt{\frac{2-\sqrt{2}}{2+\sqrt{2}}}$$

$$= \pm j \cdot 414$$

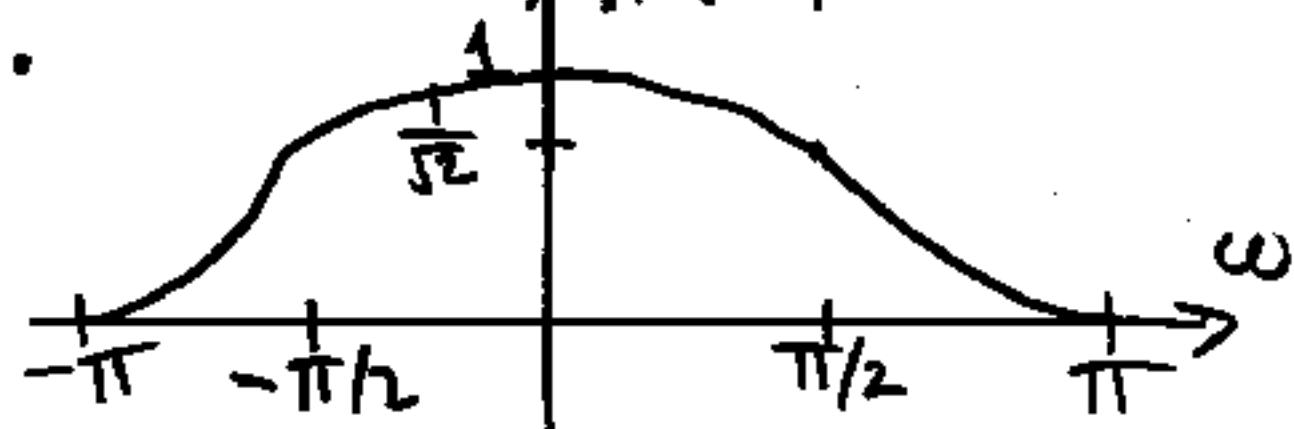


$$\begin{aligned}
 H(z) &= \frac{1 + 2z^{-1} + z^{-2}}{2 + \sqrt{2} + (2 - \sqrt{2})z^{-2}} \\
 &= Y(z) / X(z)
 \end{aligned}$$

$$y[n] = -\frac{(2-\sqrt{2})}{2+\sqrt{2}} y[n-2]$$

$$+ \frac{1}{2+\sqrt{2}} \left\{ x[n] + 2x[n-1] + x[n-2] \right\}$$

$|H(\omega)|$



Preamble to Demo  
on Using Bilinear  
Transform to Design  
Digital IIR Filters

• matlab command:

ellip.m ; ellipeg.m

Inputs: Filter order (no. of poles)

$\omega_p, \delta_1, \omega_s, \delta_2$

Outputs: coefficients of

diff. egn. for digital filter

• ellipord.m  $\Rightarrow$  to determine  
filter order a-priori

- to illustrate bilinear transform design, demo uses 's' option in ellip.m to design analog elliptic filter from pre-warped passband and stopband edges
- $$[\beta, \alpha] = \text{ellip}(\Omega_p, d_1, \Omega_s, d_2, N, 's')$$

yields

$$H_a(s) = \frac{\sum_{k=0}^{M_a} \beta_k s^k}{\sum_{k=0}^N \alpha_k s^k} \quad \boxed{\alpha_0 = 1}$$

to illustrate bilinear transform:

$$H(\omega) = H_a(s) \Big|_{s=c \frac{z-1}{z+1}} \Big|_{z=e^{j\omega}}$$

$$H(\omega) = \frac{\sum_{k=0}^{N_a} B_k \left( c \frac{e^{j\omega} - 1}{e^{j\omega} + 1} \right)^k}{\sum_{k=0}^N a_k \left( c \frac{e^{j\omega} - 1}{e^{j\omega} + 1} \right)^k}$$

- demo uses  $c = 1$
- actual analog specs:  
 $F_p$ ,  $d_1 > F_{sp}$ ,  $d_2$ , and  $F_s$   
sampling rate

• digital specs:

$$\omega_p = 2\pi \frac{F_p}{F_s}, \delta_1, \omega_s = 2\pi \frac{F_{sp}}{F_s}, \delta_2$$

• bilinear transform requires  
frequency pre-warping

$$\Omega_p = \tan\left(\frac{\omega_p}{2}\right) \quad (\neq 2\pi F_p)$$

$$\Omega_s = \tan\left(\frac{\omega_s}{2}\right) \quad (\neq 2\pi F_{sp})$$

- again: ellip.m can output digital filter design directly

⇒ don't use 's' option

- on  
 $[A, B] = \text{ellip}(N, \omega_p, \delta_1, \omega_s, \delta_2)$   
↑↑ contain filter coefficients

- Can use freqz.m to plot freq. response of digital filter given the coeffs. of diff. eqn.

- Prob. 8.16

$$F_p = 4 \text{ kHz} \quad F_{sp} = 6 \text{ kHz} \quad F_s = \\ \delta_1 = 1 \text{ dB} \quad d_2 > 40 \text{ dB} \quad 24 \text{ kHz}$$