EELE 477 Digital Signal Processing 10b z-Transforms

Inverse and Deconvolution

- We have seen that two systems in cascade can be combined into a single system by multiplying H₁(z) and H₂(z).
- We can also take a system function H(z) and *factor* it into two or more low-order systems.
- Question: can we divide the system output by the system function ("deconvolve") and recover the input?

$$Y(z) = H_1(z)X(z); \quad Y(z)H_2(z) = X(z)$$

 $Y(z) = H_1(z)H_2(z)Y(z) \Longrightarrow H_1(z)H_2(z) = 1?$

Inverse and Deconvolve, cont.

- If we can find H₂(z), it is called the inverse of H₁(z).
- NOTE that H₂(z) will not be FIR if H₁(z) is FIR.
- H₂(z) may represent a non-causal and/or unstable system even if H₁(z) is causal and stable.

Relating H(z) and H($e^{j\omega}$)

 NOTE CAREFULLY: z-transform and frequency response formulae are of identical form.

$$H(\hat{\omega}) = \sum_{k=0}^{M} b_{k} e^{-j\hat{\omega}k} \quad H(z) = \sum_{k=0}^{M} b_{k} z^{-k}$$

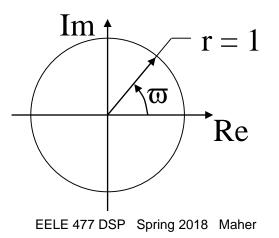
• If we evaluate H(z) for $z=e^{j\omega}$, it is clear:

$$H(\hat{\omega}) = H(e^{j\hat{\omega}}) = H(z)\Big|_{z=e^{j\omega}}$$

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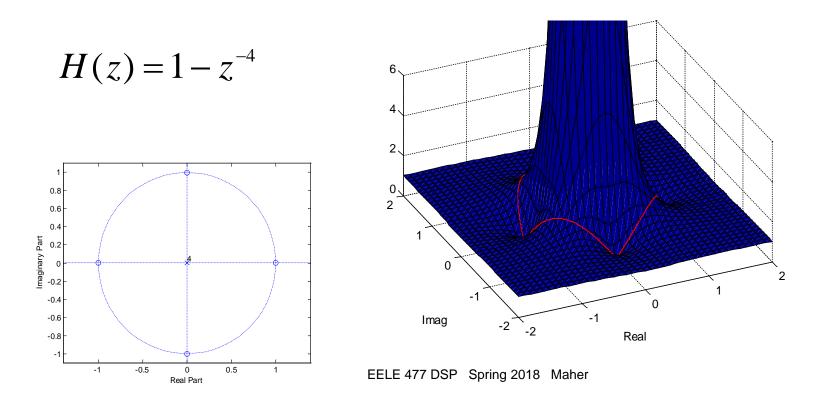
Properties of z=e^j^ω

- Observe z=e^{j∞} for -π< ω <π: |z|=1, phase= ω
- This defines a *circle* in the z-plane with radius=1: referred to as the *unit circle*



Visualizing Frequency Response

• We can observe z-transform along the unit circle to reveal the frequency response.



Poles and Zeros

- A *pole* in the z-domain is a value of *z* that "pushes up" the magnitude like a tent pole.
- A *zero* in the z-domain is a value of *z* that "pins down" the magnitude like a stake or tack.
- The pole and zero locations control the magnitude everywhere, *including along the unit circle*.

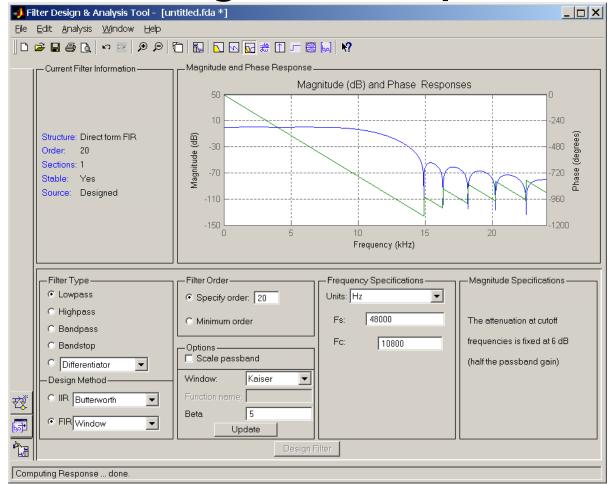
FIR Systems

- FIR systems contain only finite zeros.
 Poles are located at zero (and perhaps infinity).
- FIR filter design requires a careful choice of zero locations.
- Stop band has zeros on the unit circle.
- Pass band has zeros off the unit circle.

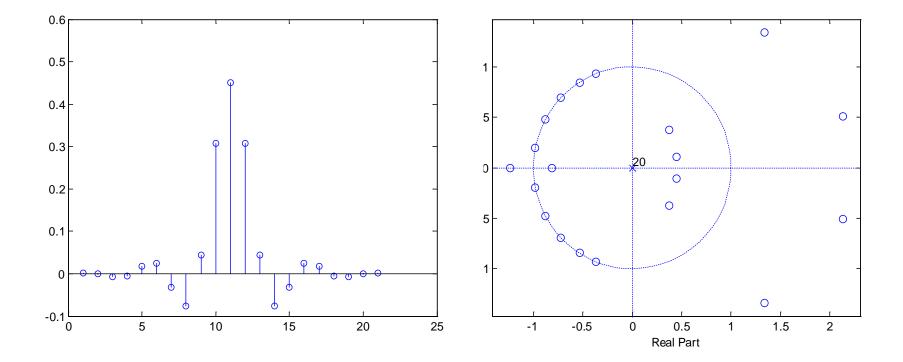
Matlab FIR Filter Design

- Matlab provides several FIR filter design tools, including: fir1, fir2, and remez
- Matlab GUI: Filter Design and Analysis Tool (FDATool)
- Usually specify passband ripple, stopband attenuation, band edges, filter order, and f_s

Design Example



Design Example (cont.)



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Symmetry and Linear Phase

- FIR systems with symmetric coefficients (*b_k=b_{M-k}*) have frequency responses with *linear phase*.
- Show this by grouping z-transform terms, for example:

$$H(z) = b_0 + b_1 z^{-1} + b_2 z^{-2} + b_3 z^{-3} + b_2 z^{-4} + b_1 z^{-5} + b_0 z^{-6}$$

= $z^{-3} [b_0 (z^3 + z^{-3}) + b_1 (z^2 + z^{-2}) + b_2 (z^1 + z^{-1}) + b_3]$

Linear Phase (cont.)

• Now evaluate H(z) on unit circle:

$$H(e^{j\hat{\omega}}) = \underbrace{e^{-j3\hat{\omega}}}_{\substack{\text{linear}\\\text{phase}\\\text{term}}} \left[\underbrace{b_0 \left(e^{j3\hat{\omega}} + e^{-j3\hat{\omega}} \right)}_{2\cos(3\hat{\omega})} + b_1 \left(e^{j2\hat{\omega}} + e^{-j2\hat{\omega}} \right)}_{2\cos(2\hat{\omega})} + b_2 \left(e^{j\hat{\omega}} + e^{-j\hat{\omega}} \right)}_{2\cos(\hat{\omega})} + b_3 \right]$$

a real function of $\hat{\omega}$ (phase=0)

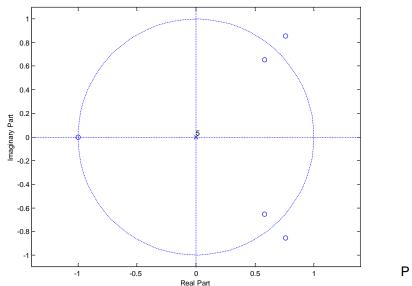
• Example if M is odd:

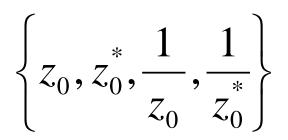
$$H(z) = b_0 + b_1 z^{-1} + b_2 z^{-2} + b_2 z^{-3} + b_1 z^{-4} + b_0 z^{-5}$$

= $z^{-2.5} \Big[b_0 \Big(z^{2.5} + z^{-2.5} \Big) + b_1 \Big(z^{1.5} + z^{-1.5} \Big) + b_2 \Big(z^{0.5} + z^{-0.5} \Big) \Big]$

Zero Symmetry

For an FIR linear phase system (implies coefficient symmetry), the zeros will have a specific pattern. For each z₀, there will be:





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