

GCT535: Sound Technology for Multimedia

# Fundamentals of DSP - Part 2



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# Goals

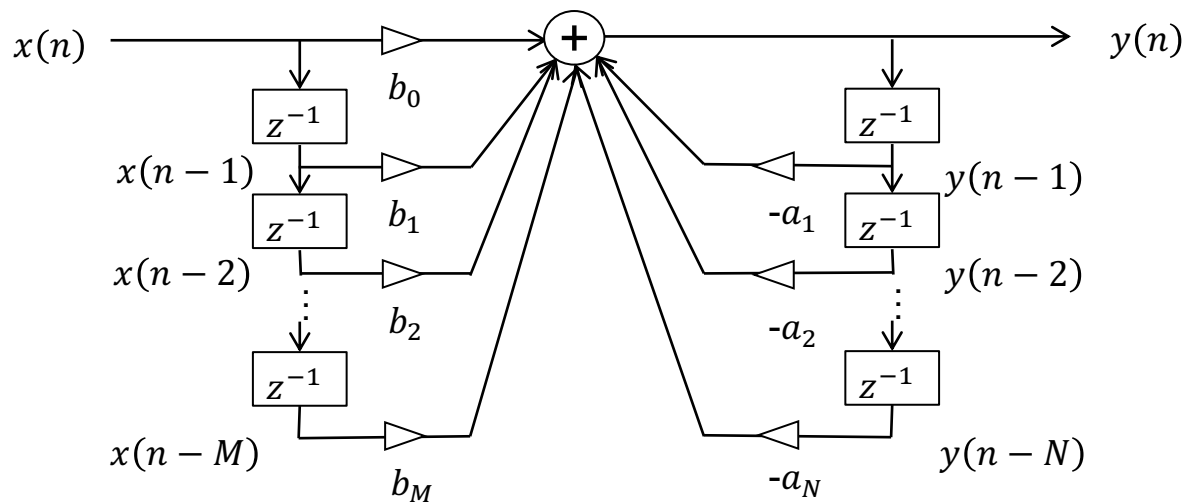
- Understanding frequency response using Z-transform and pole-zero analysis

# General Form of LTI Systems

- Difference equation

- $$y(n) = b_0 \cdot x(n) + b_1 \cdot x(n - 1) + b_2 \cdot x(n - 2) + \dots + b_M \cdot x(n - M) - a_1 \cdot y(n - 1) - a_2 \cdot y(n - 2) - \dots - a_N \cdot y(n - N)$$

- Signal flow graph



# Frequency Response

- Sine-wave analysis

- $x(n) = e^{j\omega n} \rightarrow x(n - m) = e^{j\omega(n-m)} = e^{-j\omega m} x(n)$  for any  $m$
- $y(n) = G(\omega)e^{j(\omega n + \theta(\omega))} \rightarrow y(n - m) = e^{-j\omega m} y(n)$  for any  $m$

- Putting this into the different equation

$$y(n) = b_0 \cdot x(n) + b_1 \cdot e^{-j\omega} \cdot x(n) + b_2 \cdot e^{-j2\omega} \cdot x(n) + \dots + b_M \cdot e^{-j\omega M} \cdot x(n) \\ - a_1 \cdot e^{-j\omega} \cdot y(n) - a_2 \cdot e^{-j2\omega} \cdot y(n) - \dots - a_N \cdot e^{-j\omega N} \cdot y(n)$$

$$y(n) = \frac{b_0 + b_1 \cdot e^{-j\omega} + b_2 \cdot e^{-j2\omega} + \dots + b_M \cdot e^{-j\omega M}}{1 + a_1 \cdot e^{-j\omega} + a_2 \cdot e^{-j2\omega} + \dots + a_N \cdot e^{-j\omega N}} x(n)$$

$$H(\omega) = \frac{b_0 + b_1 \cdot e^{-j\omega} + b_2 \cdot e^{-j2\omega} + \dots + b_M \cdot e^{-j\omega M}}{1 + a_1 \cdot e^{-j\omega} + a_2 \cdot e^{-j2\omega} + \dots + a_N \cdot e^{-j\omega N}}$$

$H(\omega)$  : frequency response

$G(\omega) = |H(\omega)|$  : amplitude response

$\theta(\omega) = \angle H(\omega)$  : phase response

# Frequency Response

- In general, it is very complicated to obtain analytic expressions of the amplitude and phase responses from the frequency response

- For example,  $y(n) = x(n) + 0.9 \cdot y(n - 1)$

$$H(\omega) = \frac{1}{1 - 0.9 \cdot e^{-j\omega}} \quad \rightarrow \quad G(\omega), \theta(\omega) ?$$

# Z-Transform

- Z-transform

- Define  $z$  as a variable in the complex plane: we call it  $z$ -plane
- When  $z = e^{j\omega}$  (on unit circle), the frequency response is a particular case of the following form

$$H(z) = \frac{B(z)}{A(z)} = \frac{b_0 + b_1 \cdot z^{-1} + b_2 \cdot z^{-2} + \dots + b_M \cdot z^{-M}}{1 + a_1 \cdot z^{-1} + a_2 \cdot z^{-2} + \dots + a_N \cdot z^{-N}}$$

- We call this  **$z$ -transform** or **transfer function** of the filter
- “ $z^{-1}$ ” corresponds to one sample delay: delay operator or delay element
- Digital systems are often expressed as  $z$ -transform: polynomial of  $z^{-1}$ 
  - FIR:  $H(z) = 1 + z^{-1}$ ,  $H(z) = 1 - z^{-1}$
  - IIR:  $H(z) = \frac{1}{1-0.9 \cdot z^{-1}}$ ,  $H(z) = \frac{1}{1-0.9 \cdot z^{-1} + 0.81 \cdot z^{-2}}$

# Z-Transform

- More formally, z-transform is defined as:

$$X(z) = \sum_{n=-\infty}^{\infty} x(n) \cdot z^{-n}$$

- With delay by  $M$

$$X(z)z^{-M} = \sum_{n=-\infty}^{\infty} x(n - M) \cdot z^{-n} \quad (\text{Shift Theorem})$$

# Z-Transform

- If this is applied to the difference equation of general form of LTI systems

$$Y(z) = b_0 \cdot X(z) + b_1 \cdot z^{-1}X(z) + b_2 \cdot z^{-2}X(z) + \dots + b_M \cdot z^{-M}X(z) \\ - a_1 \cdot z^{-1}Y(z) - a_2 \cdot z^{-2}Y(z) - \dots - a_N \cdot z^{-N}Y(z)$$

$$\frac{Y(z)}{X(z)} = \frac{b_0 + b_1 \cdot z^{-1} + b_2 \cdot z^{-2} + \dots + b_M \cdot z^{-M}}{1 + a_1 \cdot z^{-1} + a_2 \cdot z^{-2} + \dots + a_N \cdot z^{-N}} = H(z)$$

$$Y(z) = H(z)X(z) \quad (\text{Convolution Theorem})$$



# Z-Transform

- By replacing  $z$  with  $e^{j\omega}$  back, z-transform is represented with

$$H(e^{j\omega}) = \sum_{n=-\infty}^{\infty} h(n) \cdot e^{-j\omega n}$$

- Note that the the frequency response is discrete-time Fourier transform of impulse response
  - Also, note that the convolution theorem works for discrete-time Fourier transform or discrete Fourier transform

$$Y(e^{j\omega}) = H(e^{j\omega})X(e^{j\omega})$$

$$Y(k) = H(k)X(k)$$

# Poles and Zeros in Z-Transform

- The polynomial of  $z^{-1}$  in  $H(z)$  can be factorized
  - We can find roots for both numerator and denominator
  - Zeros: roots of numerator
  - Poles: roots of denominator

$$H(z) = \frac{B(z)}{A(z)} = \frac{(1 - q_1 z^{-1})(1 - q_2 z^{-1})(1 - q_3 z^{-1}) \dots (1 - q_M z^{-1})}{(1 - p_1 z^{-1})(1 - p_2 z^{-1})(1 - p_3 z^{-1}) \dots (1 - p_N z^{-1})}$$

- We can analyze frequency response more easily using poles and zeros than numerator or denominator coefficient

# Pole-Zero Analysis: Amplitude Response

- Amplitude Response
  - Computed using distances between poles and unit circles and distances between zeros and units circles on Z-plane

$$G(\omega) = |H(\omega)| = \left| \frac{B(\omega)}{A(\omega)} \right| = \left| \frac{(1 - q_1 e^{-j\omega})(1 - q_2 e^{-j\omega})(1 - q_3 e^{-j\omega}) \dots (1 - q_M e^{-j\omega})}{(1 - p_1 e^{-j\omega})(1 - p_2 e^{-j\omega})(1 - p_3 e^{-j\omega}) \dots (1 - p_M e^{-j\omega})} \right|$$

$$|H(\omega)| = \left| \frac{B(\omega)}{A(\omega)} \right| = \left| \frac{(e^{j\omega} - q_1)(e^{j\omega} - q_2)(e^{j\omega} - q_3) \dots (e^{j\omega} - q_M)}{(e^{j\omega} - p_1)(e^{j\omega} - p_2)(e^{j\omega} - p_3) \dots (e^{j\omega} - p_M)} \right|$$

$$|H(\omega)| = \left| \frac{B(\omega)}{A(\omega)} \right| = \frac{|(e^{j\omega} - q_1)| |(e^{j\omega} - q_2)| |(e^{j\omega} - q_3)| \dots |(e^{j\omega} - q_M)|}{|(e^{j\omega} - p_1)| |(e^{j\omega} - p_2)| |(e^{j\omega} - p_3)| \dots |(e^{j\omega} - p_M)|}$$

# Example: Reson Filter

- Difference equation

- $y(n) = x(n) + 2r \cdot \cos\theta \cdot y(n-1) - r^2 \cdot y(n-2) - x(n-2)$

- Transfer function

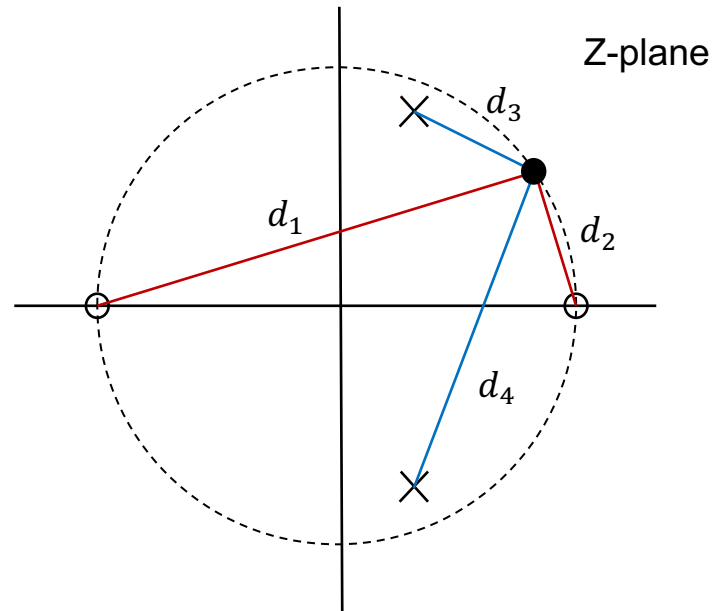
- $H(z) = \frac{1-z^{-2}}{1-2r\cos\theta \cdot z^{-1}+r^2 \cdot z^{-2}}$

- Poles:  $r(\cos\theta + j\sin\theta), r(\cos\theta - j\sin\theta)$

- Zeros: 1, -1

- Amplitude response

- $G(\omega) = |H(\omega)| = \left| \frac{B(\omega)}{A(\omega)} \right| = \frac{d_1(\omega)d_2(\omega)}{d_3(\omega)d_4(\omega)}$



# Poles and Stability

- If Poles are inside the unit circle
  - The system is stable
  - $r < 1$ : the impulse response decays with oscillation
- If Poles are on the unit circle
  - The system is critically-stable
  - $r = 1$ : the impulse response oscillates with constant amplitude (sine generation)
- If Poles are outside the unit circle
  - The system is unstable (e.g. howling when mics are closely placed near speakers)
  - $r > 1$ : the impulse response diverges

# Pole-Zero Analysis: Phase Response

- Phase Response

- Computed using angles between poles and unit circles and angles between zeros and unit circles on Z-plane

$$\begin{aligned}\theta(\omega) = \angle H(\omega) &= \frac{\angle B(\omega)}{\angle A(\omega)} = \frac{\angle(1 - q_1 e^{-j\omega})(1 - q_2 e^{-j\omega})(1 - q_3 e^{-j\omega}) \dots (1 - q_M e^{-j\omega})}{\angle(1 - p_1 e^{-j\omega})(1 - p_2 e^{-j\omega})(1 - p_3 e^{-j\omega}) \dots (1 - p_M e^{-j\omega})} \\ &= \angle(1 - q_1 e^{-j\omega}) + \angle(1 - q_2 e^{-j\omega}) + \angle(1 - q_3 e^{-j\omega}) + \dots + \angle(1 - q_M e^{-j\omega}) \\ &\quad - \angle(1 - p_1 e^{-j\omega}) - \angle(1 - p_2 e^{-j\omega}) - \angle(1 - p_3 e^{-j\omega}) - \dots - \angle(1 - p_M e^{-j\omega})\end{aligned}$$

$$\begin{aligned}\theta(\omega) = \angle H(\omega) &= \frac{\angle B(\omega)}{\angle A(\omega)} = \angle(e^{j\omega} - q_1) + \angle(e^{j\omega} - q_2) + \angle(e^{j\omega} - q_3) + \dots + \angle(e^{j\omega} - q_M) \\ &\quad - \angle(e^{j\omega} - p_1) - \angle(e^{j\omega} - p_2) - \angle(e^{j\omega} - p_3) - \dots - \angle(e^{j\omega} - p_M)\end{aligned}$$

# Example: Reson Filter

- Difference equation

- $y(n) = x(n) + 2r \cdot \cos\theta \cdot y(n-1) - r^2 \cdot y(n-2) - x(n-2)$

- Transfer function

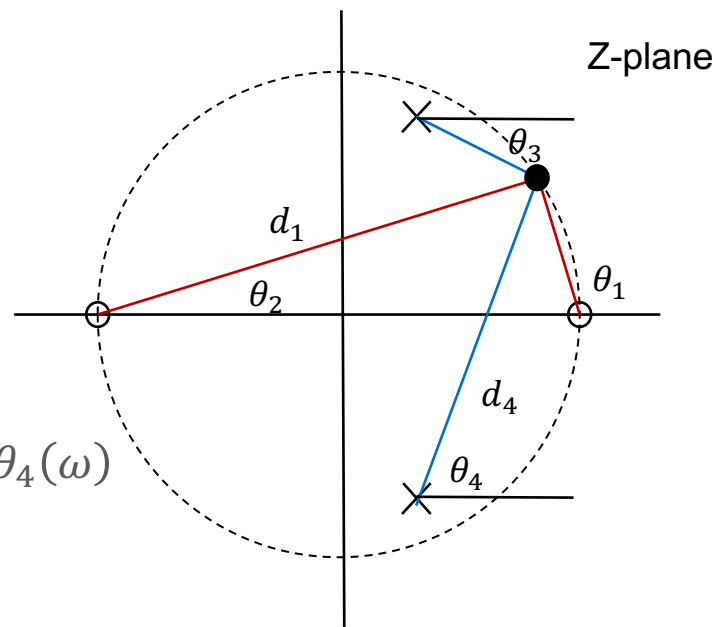
- $H(z) = \frac{1-z^{-2}}{1-2r\cos\theta \cdot z^{-1}+r^2 \cdot z^{-2}}$

- Poles:  $r(\cos\theta + j\sin\theta), r(\cos\theta - j\sin\theta)$

- Zeros: 1, -1

- Phase response

- $\angle H(\omega) = \frac{\angle B(\omega)}{\angle A(\omega)} = \theta_1(\omega) + \theta_2(\omega) - \theta_3(\omega) - \theta_4(\omega)$



# Practical Filters

- **One-pole one-zero filters**

- Leaky integrator
- DC-removal filters
- Bass / treble shelving filter (EQ)

$$H(z) = \frac{B(z)}{A(z)} = \frac{b_0 + b_1 \cdot z^{-1}}{1 + a_1 \cdot z^{-1}}$$

- **Biquad filters**

- Reson filter
- Band-pass / notch filters
- Equalizers

$$H(z) = \frac{B(z)}{A(z)} = \frac{b_0 + b_1 \cdot z^{-1} + b_2 \cdot z^{-2}}{1 + a_1 \cdot z^{-1} + a_2 \cdot z^{-2}}$$

- **Any high-order filter** can be factored into a **cascade of one-pole one-zero filters** or **bi-quad filters**

$$H(z) = H_o(z) \prod H_{bi}(z)$$



# Digital Audio Effects

- Filters, Equalizers
  - One-pole/one-zero filter and bi-quad filters
  - Small number of coefficients and small size of delay elements
- Delay-based Effects
  - Chorus, Flanger
  - FIR or IIR filters
  - Small number of coefficients and a mid-to-long size of delay lengths
- Spatial Effects
  - HRTF, Reverb
  - FIR or IIR filters
  - A large number of coefficients and long delay lengths