Lecture 5 Frequency Domain Representation of Discrete Time Systems

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Signal & Frequency Components

- Continuous periodic signal Discrete frequency components (Fourier Series)
- Continuous non-periodic signal Continuous frequency components (CTFT)
- Discrete-time signal components (DTFT)

Linear Combination

 When a signal can be represented as a linear combination of complex exponentials:

$$x[n] = \sum_{k} a_k e^{j\omega_k n}$$

knowing the response of the LTI system to a single sinusoidal signal, we can determine its response to more complicated signals by making use of superposition property.

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$$x[n] \longrightarrow h[n] \qquad y[n]$$

$$x[n] = \delta[n] \longrightarrow h[n]$$

$$x[n] = e^{j\omega n} \qquad h[n] \longrightarrow y[n]$$

$$y[n] = h[n] \bigoplus e^{j\omega n} = \sum_{k=-\infty}^{\infty} h[k] e^{j\omega(n-k)}$$

$$= \left(\sum_{k=-\infty}^{\infty} h[k] e^{-j\omega k}\right) e^{j\omega n}$$

Let

$$H(e^{j\omega}) = \sum_{k=-\infty}^{\infty} h[k]e^{-j\omega k}$$

Eigenfunction

Then, we can write

$$y[n] = H(e^{j\omega})e^{j\omega n}$$

- Thus, for a complex exponential input signal $e^{j\omega n}$, the output of an LTI discrete-time system is also a complex exponential signal of the same frequency multiplied by a complex constant $H(e^{j\omega})$.
- If applying a function as an input to a system, and the output of the system is the same function multiplied by a constant, such function is an **eigenfunction** of the system.
- So, $e^{j\omega n}$, is an eigenfunction of the system.

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The Frequency Response

$$H(e^{j\omega}) = \sum_{k=-\infty}^{\infty} h[k]e^{-j\omega k}$$

is defined to be the **frequency response** of the LTI system with impulse response h[n]

- $H(e^{j\omega}) = H_{\rm re}(e^{j\omega}) + jH_{\rm im}(e^{j\omega})$
- $H(e^{j\omega}) = |H(e^{j\omega})|e^{j\theta(\omega)}$, where, $\theta(\omega) = \arg\{H(e^{j\omega})\}$
- $|H(e^{j\omega})|$: magnitude response
- $\theta(\omega)$: phase response

Example

- Consider the ideal delay system defined by $y[n] = x[n n_d]$, for constant integer n_d
- With input $x[n] = e^{j\omega n}$, we have $y[n] = e^{j\omega(n-n_d)} = e^{-j\omega n_d}e^{j\omega n}$

The frequency response of the ideal delay is therefore

$$H(e^{j\omega}) = e^{-j\omega n_d}$$

• An alternative method: the impulse response of the ideal delay is $h[n] = \delta[n - n_d]$. So the frequency response of the ideal delay system is

$$H(e^{j\omega}) = \sum_{n=-\infty}^{\infty} \delta[n - n_d] e^{-j\omega n} = e^{-j\omega n_d}$$

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Frequency Response in Decibels

• Gain Function:

$$\mathcal{G}(\omega) = 20\log_{10}\left|H(e^{j\omega})\right|$$

the unit is in dB

Attenuation (or loss function):

$$\mathcal{A}(\omega) = -20\log_{10}\left|H(e^{j\omega})\right|$$

is the negative of the gain function.

Symmetry of frequency Response

- Due to DTFT, for a real impulse response h[n], $H(e^{j\omega})$ is conjugate symmetric, i.e.,
 - $H(e^{j\omega}) = H^*(e^{-j\omega})$, or
 - $|H(e^{j\omega})| = |H(e^{-j\omega})|$, and $\theta(\omega) = -\theta(-\omega)$, or
 - $H_{\rm re}(e^{j\omega})$ is even and $H_{\rm im}(e^{j\omega})$ is odd.
- For a real symmetric impulse response,
 - $H(e^{j\omega})$ is real and symmetric.

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Frequency-Domain Characterization of LTI DT System

• For LTI system in time domain, we have $y[n] = x[n] \oplus h[n]$

$$x[n] \qquad b[n] \qquad y[n]$$

• Applying convolution property of DTFT, we have $Y(e^{j\omega}) = X(e^{j\omega})H(e^{j\omega})$

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

Frequency Response of FIR System

The time-domain input-output relation of FIR system:

$$y[n] = \sum_{k=N_1}^{N_2} h[k]x[n-k], \quad N_1 < N_2$$

Applying DTFT on both sides, we arrive at

$$Y(e^{j\omega}) = \sum_{k=N_1}^{N_2} h[k]e^{-j\omega k}X(e^{j\omega}),$$

The frequency response of FIR system is given by

$$H(e^{j\omega}) = \sum_{k=N_1}^{N_2} h[k]e^{-j\omega k},$$

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Frequency Response of IIR System

• The time-domain input-output relation of IIR system

$$\sum_{m=0}^{N} b_m y[n-m] = \sum_{m=0}^{M} a_m x[n-m]$$

Applying DTFT on both sides, we arrive at

$$\sum_{m=0}^{N} b_m e^{-j\omega m} Y(e^{j\omega}) = \sum_{m=0}^{M} a_m e^{-j\omega m} X(e^{j\omega})$$

• The frequency response of IIR system is given by

$$H(e^{j\omega}) = \frac{Y(e^{j\omega})}{X(e^{j\omega})} = \frac{\sum_{m=0}^{M} a_m e^{-j\omega m}}{\sum_{m=0}^{N} b_m e^{-j\omega m}}$$

Example

- Determine the frequency response of the *M*-point moving average filter.
- Since the input-output relation is given by:

$$y[n] = \frac{1}{M} \sum_{l=0}^{M-1} x[n-l]$$

the impulse response is given by:

$$h[n] = \frac{1}{M} \sum_{l=0}^{M-1} \delta[n-l] = \begin{cases} \frac{1}{M}, & 0 \le n \le M-1\\ 0, & \text{otherwise} \end{cases}$$

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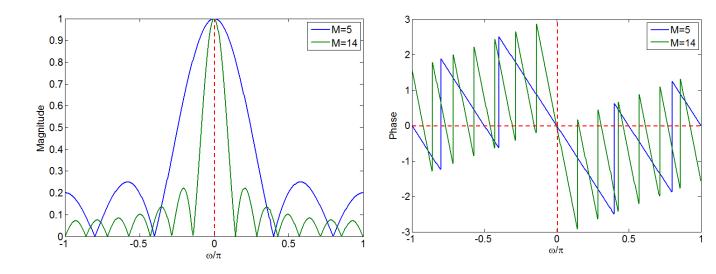
Thus, the frequency response is given by

$$H(e^{j\omega}) = \frac{1}{M} \sum_{n=0}^{M-1} e^{-j\omega n} = \frac{1}{M} \cdot \frac{1 - e^{-jM\omega}}{1 - e^{-j\omega}}$$
$$= \frac{1}{M} \cdot \frac{\sin(M\omega/2)}{\sin(\omega/2)} e^{-j(M-1)\omega/2}$$

$$|H(e^{j\omega})| = \frac{1}{M} \left| \frac{\sin\left(\frac{M\omega}{2}\right)}{\sin\left(\frac{\omega}{2}\right)} \right|$$

$$\theta(\omega) = \frac{-(M-1)\omega}{2} + \pi \sum_{k=1}^{\lfloor M/2 \rfloor} \mu\left(\omega - \frac{2\pi k}{M}\right)$$

The plots of the magnitude response and phase response of the M-point moving average filter, for M=5 and M=14

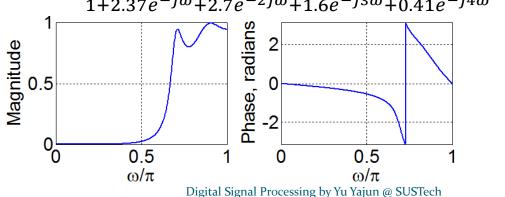


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Unwrapped Phase Function

- The principle value of phase function is defined to within a range $[-\pi, \pi]$.
- The phase function of DTFT thus computed exhibits discontinuity of 2π radians in plots.
- Example: $X(e^{j\omega}) =$ $0.008 - 0.033e^{-j\omega} + 0.05e^{-j2\omega} - 0.033e^{-j3\omega} + 0.008e^{-j4\omega}$ $1+2.37e^{-j\omega}+2.7e^{-2j\omega}+1.6e^{-j3\omega}+0.41e^{-j4\omega}$

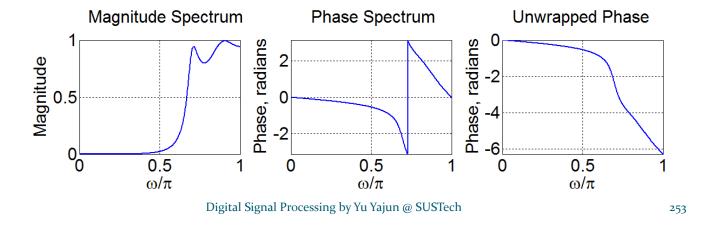


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Unwrapped Phase Function

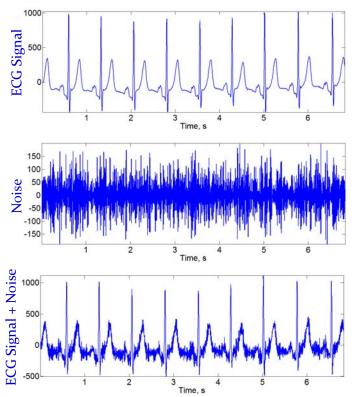
• The process to remove the 2π discontinuity is called **unwrapping the phase.**

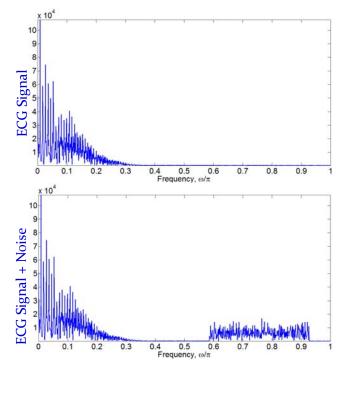


The Concept of Filtering

- One application of an LTI discrete-time system is to pass certain frequency components in an input sequence without any distortion (if possible) and to block other frequency components.
- Such systems are called digital filters and are one of main devices in digital signal processing

- A time domain signal with noise
- Their frequency spectrum





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The Concept of Filtering

• Any discrete-time signal may be expressed as

$$x[n] = \frac{1}{2\pi} \int_{-\pi}^{\pi} X(e^{j\omega}) e^{j\omega n} d\omega$$

- Any frequency component $e^{j\omega n}$ may be scaled by a frequency response $H(e^{j\omega})$ at frequency ω , such that the frequency component is passed without distortion, or attenuated.
- For example, if we have an ideal LTI system with magnitude response given by

$$\left| \frac{H(e^{j\omega})}{0} \right| = \begin{cases} 1, & |\omega| \le \omega_c \\ 0, & \omega_c < |\omega| \le \pi \end{cases}$$

A Simple Example

• We apply an input x[n] to the system, where

$$x[n] = A\cos\omega_1 n + B\cos\omega_2 n,$$

$$0 < \omega_1 < \omega_c < \omega_2 < \pi$$

Because of linearity, the output of the system is

$$y[n]$$

$$= A |H(e^{j\omega_1})| \cos(\omega_1 n + \theta(\omega_1))$$

$$+ B |H(e^{j\omega_2})| \cos(\omega_2 n + \theta(\omega_2))$$

- As $|H(e^{j\omega_1})| = 1$, and $|H(e^{j\omega_2})| = 0$, the output reduces to $y[n] = A\cos(\omega_1 + \theta(\omega_1))$
- The LTI system acts like a lowpass filter.

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Design Example

- Design a very simple digital filter.
- Requirement: An input, consisting of two sinusoidal sequences of angular frequencies 0.1 rad/sample and 0.4 rad/sample, is to be filtered to keep the high-frequency component, but block the low-frequency component.
- For simplicity, we assume a filter of length 3 with an impulse response: $h[0]=h[2]=\alpha$, and $h[1]=\beta$.

- The input-output relation in time-domain would be: y[n] = h[0]x[n] + h[1]x[n-1] + h[2]x[n-2] $= \alpha x[n] + \beta x[n-1] + \alpha x[n-2]$
- **Design objective:** Choose suitable values of α and β , such that the output contains only a sinusoidal sequence with an angular frequency 0.4 rad/sample.
- Now the frequency response of the filter is given by, $H(e^{j\omega}) = h[0] + h[1]e^{-j\omega} + h[2]e^{-j2\omega}$ $= \alpha(1 + e^{-j2\omega}) + \beta e^{-j\omega}$ $= 2\alpha \left(\frac{e^{j\omega} + e^{-j\omega}}{2}\right)e^{-j\omega} + \beta e^{-j\omega} = (2\alpha\cos\omega + \beta)e^{-j\omega}$

Magnitude responsegital Signal Processing by Yu Yajun @ SUSTech

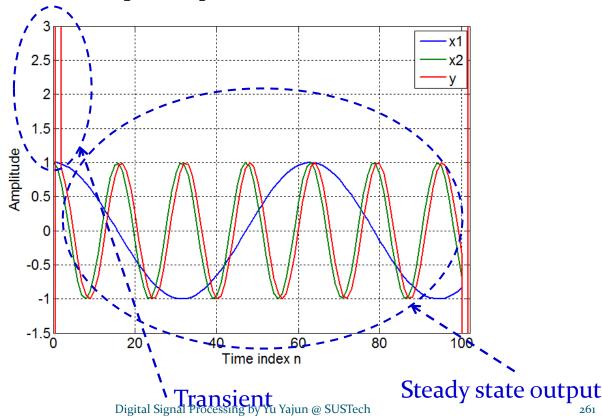
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- To block the low-frequency component, let $H(e^{j0.1}) = (2\alpha\cos(0.1) + \beta) = 0$
- To pass the high-frequency component, let $H(e^{j0.4}) = (2\alpha\cos(0.4) + \beta) = 1$
- Result in:

$$\alpha = -6.76185$$
, $\beta = 13.456335$ i.e., $h[n] = \{-6.76185, 13.456335, -6.76185\}$, for $n = 0, 1, 2$

• So the designed filter has the input-output relation in time-domain given by y[n] = -6.76185(x[n] + x[n-2]) + 13.456335x[n-1] and the input is $x[n] = (\cos(0.1n) + \cos(0.4n))\mu[n]$

Input and output sequences in time-domain



Phase Delay and Group Delay

Re-examine a system with an input of pure sinusoidal signal

$$y[n] = h[n] \bigotimes A\cos(\omega_0 n + \varphi)$$

$$= A \left(\sum_{k=-\infty}^{\infty} h[k] e^{-j\omega_0 k} \right) \cos(\omega_0 n + \varphi) \qquad (-\tau_p(\omega_0))$$

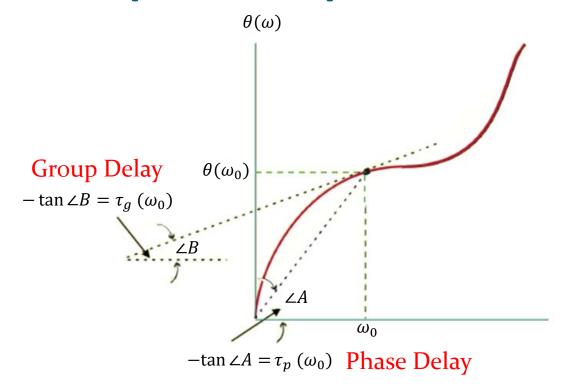
$$= A |H(e^{j\omega_0})| \cos(\omega_0 n + \theta(\omega_0) + \varphi)$$

$$= A |H(e^{j\omega_0})| \cos\left(\omega_0 \left[n + \frac{\theta(\omega_0)}{\omega_0}\right] + \varphi\right)$$

Define phase delay and group delay, respectively, as

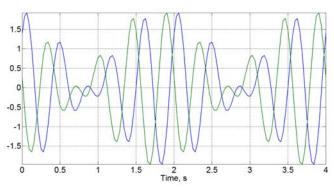
$$au_p(\omega_0) = -rac{ heta(\omega_0)}{\omega_0}, \qquad au_g(\omega) = -rac{d heta(\omega)}{d\omega}.$$

A Graphic Comparison

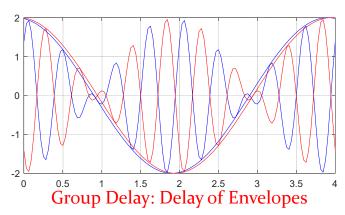


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*Physical Meanings



Phase Delay: Delay of Samples



•
$$T = \frac{1}{32}$$
, $\omega_1 = 4\pi$, $\omega_2 = 5\pi$

• Blue Signal: $\sin(\omega_1 n + 0.2\pi) + \sin(\omega_2 n + 0.3\pi)$

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• Green Signal:

$$\sin(\omega_1 n + 0.2\pi + 5\omega_1 T) + \sin(\omega_2 n + 0.3\pi + 5\omega_2 T)$$

$$\theta(\omega_1) = 5\omega_1 T, \ \theta(\omega_2) = 5\omega_2 T,$$

$$\tau_p(\omega_1) = \tau_p(\omega_2) = -5T$$

$$\tau_q(\omega_1) = \tau_q(\omega_2) = -5T$$

• Red Signal:

$$\sin(4\pi n + 0.2\pi + 5\omega_1 T + 0.4\pi) + \sin(5\pi n + 0.3\pi + 5\omega_2 T + 0.2\pi)$$

$$\theta(\omega_1) = 5\omega_1 T + 0.4\pi, \ \theta(\omega_2) = 5\omega_2 T + 0.2\pi,$$

$$\tau_p(\omega_1) = -5T - 0.1$$

$$\tau_p(\omega_2) = -5T - 0.04$$

$$\tau_g(\omega_1) = \tau_g(\omega_2) = -\frac{\Delta\theta(\omega)}{\Delta\omega} - \frac{\theta(\omega_2) - \theta(\omega_1)}{\omega_2 - \omega_1}$$

$$= -(5T - 0.2) = 1.4T$$

•
$$H(e^{j\omega}) = (2\alpha\cos\omega + \beta)e^{-j\omega}$$
,

•
$$\tau_p(\omega) = -\frac{\theta(\omega)}{\omega} = 1$$
, $\tau_g(\omega) = -\frac{d\theta(\omega)}{d\omega} = 1$

