數位訊號處理概論 Introduction to Digital Signal Processing: HW5, Quiz5 TA Review

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Remind – Final Exam

- time: **2022/06/15 13:20-15:10**
- scope: everything we cover this semester
- 採實體考試與線上考試並行
- 需要線上考試的同學請在6/12 22:00 前在eeclass 的問卷中登記, 如要實體考試則不需額外登 記, 如沒有特殊原因不能臨時改
- 詳細規則於6/11公告
- 1 A4 cheat sheet is allowed, and printed from everywhere is not accepted.
- **Don't cheat, also we won't allow any cheating behaviors. Don't try to challenge.**

Quiz5 & HW5

- We will upload the hw5 scores on Sunday
- If you have other problems:
	- Please email **both TAs**
		- 簡婉軒: wschien@gap.nthu.edu.tw
		- Shreya: [shreya@gapp.nthu.edu.tw](mailto:Shreya@gapp.nthu.edu.tw)
	- Please explain in English for HW q5-q8 and Quiz q13-q25
	- Quiz5: Due time: **2022/06/10 17:20**
	- HW5: Due time: **2022/06/13 17:20**

- Please answer the following questions: (25%) 1.
	- (a) If a system function has zeros at the origin of the z-plane then the system function is a proper rational function. Why?
	- (b) Can you obtain the z-transform of $u[n]$ at $z = 0$? If you can, what is its value? If you cannot, why not?
	- (c) What is the preferred method for obtaining the inverse z-transform of rational functions? Describe this method.
	- (d) Explain why a one-sided z-transform is able to determine response to a LCCDE with initial conditions while the two-sided z-transform cannot.
	- (e) Can any system with a rational system function be decomposed into a product of an allpass system and a minimum-phase system? If yes, then explain how?

Please refer to professor's slides and textbook

2. Determine the impulse response of the system for all possible regions of convergence. (5%)

$$
y[n] - \frac{5}{2}y[n-1] + y[n-2] = x[n-1]
$$

Solution:

$$
H(z) = \frac{Y(z)}{X(z)} = \frac{z^{-1}}{1 - \frac{5}{2}z^{-1} + z^{-2}} = \frac{\frac{2}{3}}{1 - 2z^{-1}} + \frac{-\frac{2}{3}}{1 - \frac{1}{2}z^{-1}}
$$

ROC:
$$
|z| > 2
$$

\n
$$
h[n] = \frac{2}{3} \cdot 2^n u[n] - \frac{2}{3} \cdot \left(\frac{1}{2}\right)^n u[n]
$$
\nROC: $|z| < \frac{1}{2}$
\n
$$
h[n] = -\frac{2}{3} \cdot 2^n u[-n-1] + \frac{2}{3} \cdot \left(\frac{1}{2}\right)^n u[-n-1]
$$
\nROC: $\frac{1}{2} < |z| < 2$
\n
$$
h[n] = -\frac{2}{3} \cdot 2^n u[-n-1] - \frac{2}{3} \cdot \left(\frac{1}{2}\right)^n u[n]
$$

- Consider the finite length sequence $x[n] = u[n] u[n N]$ (15%) 3.
	- (a) Determine the z-transform $X(z)$ of the sequence $x[n]$.
	- (b) Determine and plot the sequence $y[n] = x[n] * x[n]$.
	- (c) Determine the z-transform $Y(z)$ of the sequence $y[n]$.

4. Consider a stable system with input $x[n]$ and output $y[n]$. Determine its impulse response $h[n]$ if we are given that: (5%)

$$
x[n] = \left(\frac{1}{3}\right)^{|n|} \text{ and } y[n] = 2\left(\frac{1}{3}\right)^n u[n] - 2^{n+2}u[-n]
$$
\n
$$
X(z) = \frac{1}{1 - \frac{1}{3}z^{-1}} + \frac{-1}{1 - 3z^{-1}} = \frac{-\frac{8}{3}z^{-1}}{1 - \frac{10}{3}z^{-1} + z^{-2}}, \quad \text{ROC: } \frac{1}{3} < |z| < 3
$$
\n
$$
Y(z) = \frac{2}{1 - \frac{1}{3}z^{-1}} - 4 + \frac{4}{1 - 4z^{-1}}
$$
\n
$$
= \frac{2 + 8z^{-1} - \frac{16}{3}z^{-2}}{(1 - \frac{1}{3}z^{-1})(1 - 4z^{-1})}, \quad \text{ROC: } \frac{1}{3} < |z| < 4
$$
\n
$$
H(z) = \frac{Y(z)}{X(z)} = -\frac{3}{4}z - \frac{19}{8} + \frac{3}{2}z^{-1} + \frac{-\frac{5}{4}}{1 - 4z^{-1}}, \quad \text{ROC: } |z| < 4
$$
\n
$$
h[n] = -\frac{3}{4}\delta[n+1] - \frac{19}{8}\delta[n] + \frac{3}{2}\delta[n-1] + \left(\frac{5}{4}\right) \cdot 4^n u[-n-1]
$$

5. Show that the group delay of an LTI system with frequency response function $H(e^{j\omega}) =$ $H_R(\omega) + jH_I(\omega)$ can be expressed as: (10%)

$$
\tau(\omega)=\frac{H_R(\omega)G_R(\omega)+H_I(\omega)G_I(\omega)}{|H(e^{j\omega})|^2}.
$$

Where $G(e^{j\omega}) = G_R(\omega) + jG_I(\omega)$ is the DTFT of $nh[n]$.

Proof:
\n
$$
\tau_{gd}(\omega) = -\frac{d\Psi(\omega)}{d\omega}
$$
\n
$$
\Psi(\omega) = \tan^{-1}\frac{H_I(\omega)}{H_R(\omega)} + 2k\pi, \quad \Psi(\omega) \text{ is continuous}
$$
\n
$$
-\frac{d\Psi(\omega)}{d\omega} = -\frac{d}{d\omega}\left(\tan^{-1}\frac{H_I(\omega)}{H_R(\omega)} + 2k\pi\right) = \frac{\left(\frac{dH_I(\omega)}{d\omega}\right)H_R(\omega) - \left(\frac{dH_R(\omega)}{d\omega}\right)H_I(\omega)}{H_R^2(\omega) + H_I^2(\omega)}
$$
\n
$$
nh[n] \xrightarrow{\text{DTFT}} \mathbf{j} \cdot \frac{d\left[H_R(\omega) + \mathbf{j}H_I(\omega)\right]}{d\omega} = \mathbf{j}\frac{dH_R(\omega)}{d\omega} - \frac{dH_I(\omega)}{d\omega}
$$
\nHence, we have\n
$$
\frac{dH_R(\omega)}{d\omega} = G_I(\omega); \quad \frac{dH_I(\omega)}{d\omega} = -G_R(\omega)
$$
\n
$$
H_R^2(\omega) + H_I^2(\omega) = |H(\mathbf{e}^{\mathbf{j}\omega})|^2
$$
\nThus, we proved that\n
$$
\tau_{gd}(\omega) = -\frac{d\Psi(\omega)}{d\omega} = \frac{H_R(\omega)G_R(\omega) + H_I(\omega)G_I(\omega)}{|H(\mathbf{e}^{\mathbf{j}\omega})|^2}
$$

- 6. Determine location of poles and zeros in following conditions: (20%)
	- (a) We want to design a second-order IIR filter using pole-zero placement that satisfies the following requirements: (1) The magnitude response is 0 at $\omega_1 = 0$ and $\omega_3 = \pi$; (2) The maximum magnitude is 1 at $\omega_{2,4} = \pm \pi/4$; and (3) the magnitude response is approximately

 $\frac{1}{\sqrt{2}}$ at frequencies $\omega_{2,4}$ ± 0.05. Determine the location of the two poles and zeros of the

required filter and then compute its system function H(z).

Now consider the second-order notch IIR filter that satisfies the following requirements: (1) (b) The magnitude response has notches at $\omega_{1,2} = \pm 2\pi/3$; (2) The maximum magnitude

response is 1; (3) the magnitude response is approximately $\frac{1}{\sqrt{2}}$ at frequencies $\omega_{1,2} \pm 0.01$.

Using the pole-zero placement approach determine locations of two poles and two zeros of the required filter and then compute its system function H(z).

(a) Solution:

zeros:
$$
z_1 = e^{j0} = 1
$$
, $z_2 = e^{j\pi} = -1$
poles: $p_1 = re^{j\frac{\pi}{4}}$, $p_2 = re^{-j\frac{\pi}{4}}$, $r \in (0, 1)$

The system function is:

$$
H(z) = b_0 \frac{(1 - z^{-1})(1 + z^{-1})}{(1 - re^{j\frac{\pi}{4}}z^{-1})(1 - re^{-j\frac{\pi}{4}}z^{-1})}
$$

The frequency response is:

$$
H(e^{j\omega}) = b_0 \frac{(1 - e^{-j\omega})(1 + e^{-j\omega})}{(1 - re^{j\frac{\pi}{4}}e^{-j\omega})(1 - re^{-j\frac{\pi}{4}}e^{-j\omega})}
$$

$$
Constrain |H(e^{j\omega})|_{\text{max}} = 1
$$
, we have

$$
|b_0| = \frac{(1-r)\sqrt{1+r^2}}{\sqrt{2}}
$$

Choose
$$
r = 0.95
$$
.

(b) Solution:

$$
\begin{array}{rcl}\n\text{zeros:} & z_1 = e^{j\frac{2\pi}{3}}, \quad z_2 = e^{-j\frac{2\pi}{3}} \\
\text{poles:} & p_1 = r e^{j\left(\frac{2\pi}{3} + \phi\right)}, \quad p_2 = r e^{-j\left(\frac{2\pi}{3} + \phi\right)}, \quad r \in (0, 1)\n\end{array}
$$

The system function is:

$$
H(z) = b_0 \frac{(1 - e^{j\frac{2\pi}{3}}z^{-1})(1 - e^{-j\frac{2\pi}{3}}z^{-1})}{(1 - re^{j(\frac{2\pi}{3} + \phi)}z^{-1})(1 - re^{-j(\frac{2\pi}{3} + \phi)}z^{-1})}
$$

The frequency response is:

$$
H(e^{j\omega}) = b_0 \frac{(1 - e^{j\frac{2\pi}{3}}e^{-j\omega})(1 - e^{-j\frac{2\pi}{3}}e^{-j\omega})}{(1 - re^{j(\frac{2\pi}{3} + \phi)}e^{-j\omega})(1 - re^{-j(\frac{2\pi}{3} + \phi)}e^{-j\omega})}
$$

Constrain $|H(e^{j\omega})|_{\text{max}} = 1$, we have

$$
|b_0| = \frac{(1-r)|1 - re^{-2j(\frac{2\pi}{3} + \phi)}}{|1 - e^{-j\phi}||1 - e^{-j(\frac{4\pi}{3} + \phi)}|}
$$

Choose $r = 0.9, \phi = 0.01$.

7. Derive (eq-q7.1) formula for the impulse response of an ideal bandpass filter by (10%)

$$
h_{bp}[n] = 2 \frac{\sin \omega_c (n - n_d)}{\pi (n - n_d)} \cos \omega_0 n
$$

Using impulse response (eq-q7.2) of the ideal low pass filter and the modulation property of DTFT.

$$
h_{lp}[n] = \frac{\sin \omega_c (n - n_d)}{\pi (n - n_d)}
$$
 eq-q7.2
Solution:

$$
h_{bp}[n] = 2 \frac{\sin \omega_c (n - n_d)}{\pi (n - n_d)} \cos \omega_0 n \quad (5.72)
$$

$$
h_{lp}[n] = \frac{\sin \omega_c (n - n_d)}{\pi (n - n_d)} \quad (5.70)
$$

Modulation Property:

$$
x[n] \cos \omega_0 n = \frac{1}{2} X(e^{j(\omega + \omega_0)}) + \frac{1}{2} X(e^{j(\omega - \omega_0)})
$$

$$
H_{bp}(e^{j\omega}) = \begin{cases} e^{-j\omega n_d}, & \omega_{\ell} \le |\omega| \le \omega_h \\ 0, & \text{otherwise} \end{cases}
$$

$$
H_{lp}(e^{j\omega}) = \begin{cases} e^{-j\omega n_d}, & |\omega| \le \omega_c \\ 0, & \omega_c < |\omega| \le \pi \end{cases}
$$

$$
H_{bp}(e^{j\omega}) = H_{lp}(e^{j(\omega - \omega_0)}) + H_{lp}(e^{j(\omega + \omega_0)})
$$
where $\omega_0 = \frac{\omega_{\ell} + \omega_h}{2}$ and $\omega_c = \frac{\omega_h - \omega_{\ell}}{2}$.
Hence,
$$
h_{bp}[n] = 2h_{lp}\cos \omega_0 n = 2\frac{\sin \omega_c(n - n_d)}{\pi(n - n_d)}\cos \omega_0 n
$$

- 8. Determine the group delay of the following systems: (10%)
	- (a) $y[n] = x[n] 0.9x[n-1]$
	- (b) $y[n] = 0.8y[n-1] + x[n]$

 (a) Solution: The frequency response is:

$$
H(e^{j\omega}) = 1 - 0.9e^{-j\omega} = 1 - 0.9\cos\omega + j0.9\sin\omega
$$

The phase response is:

$$
\Psi(\omega) = \tan^{-1} \frac{0.9 \sin \omega}{1 - 0.9 \cos \omega} + 2k\pi
$$

The group delay is:

$$
\tau_{\text{gd}}(\omega) = -\frac{\mathrm{d}\Psi(\omega)}{\mathrm{d}\omega} = \frac{1 - 0.9\cos\omega}{1 + 0.9^2 - 1.8\cos\omega}
$$

(b) Solution:

The frequency response is:

$$
H(e^{j\omega}) = \frac{1}{1 - 0.8e^{-j\omega}} = \frac{1}{1 - 0.8\cos\omega + j0.8\sin\omega}
$$

The phase response is:

$$
\Psi(\omega) = -\tan^{-1}\frac{0.8\sin\omega}{1 - 0.8\cos\omega} + 2k\pi
$$

The group delay is:

$$
\tau_{\rm gd}(\omega)=-\frac{{\rm d}\Psi(\omega)}{{\rm d}\omega}=\frac{0.8\cos\omega-0.8^2}{1+0.8^2-1.6\cos\omega}
$$

Which of the following statement is *false* about z-transform? $\mathbf{2}$

A. Any sequence x[n] can be uniquely characterized by its z-transform. 0

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- B. One-sided z-transform is unique for causal signal.
- C. The z-transform converts convolution equations and LCCDE into algebraic equations. \circ
- D. The z-transform of a causal periodic signal can be determined from the knowledge of the z-transform of its first cycle \circ
- E. Two-sided z-transform is able to determine response to a LCCDE with initial conditions. \odot

(e) You had proved in hw5-1

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B. A digital resonator is a special two pole bandpass filter with the pair of complex conjugate poles located near the unit circle. \circ

C. A comb filter is a special case of notch filter in which the nulls occur periodically across the frequency band. \bigcirc

D. All ideal filters are unstable and unrealizable, but take ideal low pass filter as prototype filter. \circ

E. Notch filters have perfect null at certain frequencies. \circ

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難易度:未設定

- 5 Which of the following statement is *false* about group delay?
	- A. Group delay is useful in checking the linearity of the phase response.
	- B. Group delay is the time delay that the signal component of frequency w undergoes as it passes from the input to the output of the system.
	- C. If the phase $\Theta(\omega)$ of the system is linear, then the group delay of the system decreases with frequency of signal. $^{\circ}$
	- D. In the IIR filter, the group delay response is used as a measure of how much dispersal a typical input will undergo when being procured by a filter. \bigcirc
	- E. Group delay is the time delay of the amplitude envelopes of the various sinusoidal components of a signal.

(c) We know that the group delay of the system with phase $\Theta(\omega)$ is defined as $Tg(\omega)=d\Theta(\omega)d\omega$. Given the phase is linear=> the group delay of the system is constant.

- . Impulse Response is of finite duration
- . It is a non-recursive system because there is no feedback path present in the FIR filter.
- They are always be designed as a linear phase.
- · Stability is quaranteed is the FIR system.
- In FIR, to obtain the same frequency response as IIR large number of additions & multiplication is reviewed, so the speed is very slow.

Properties of IIR filter:

- The impulse response is of infinite duration.
- . IIR system is also known as a recursive system because there is a feedback path from output to input.
- · IIR system cannot be designed as a linear phase system.
- · Stability cannot be quaranteed.
- In the IIR system, fewer multiplications and additions are reviewed, so processing speed is very fast.

Linear phase:

- FIR can be easily designed to have a exact linear phase
- . No phase distortion is introduced into the signals to be filtered.
- All frequencies are shifted in time by the same amount.
- Increasing the order of the filter, but keeping all else the same, increases the sharpness of the filter roll-off
- . The sharpness of the FIR and IIR filters is very different for the same order.
- Because of the recursive nature of an IIR filter, where part of the filter output is used as an input, IIR filters have sharper roll-off with the same order FIR filter.

Which one of the following is the advantage of FIR filter over IIR filter?

A. FIR filter is always unstable.

- B. FIR filter can have an exact linear phase. \odot
- C. For FIR filter, the design methods are non-linear.
- D. FIR filter cannot be realized efficiently in hardware.
- E. In FIR system, fewer multiplications and additions are revied, so processing speed is very fast.

(c) There are always some distortions, even in the perfect transmission line. This is due to the variation of the secondary parameters. The attenuation constant causes the frequency distortion, whereas the phase constant causes the phase distortion.

- Which of the following statement is *false* about distortion?
	- A. Signal distortion is the disadvantage of sampling rate conversion by converting the signal into analog signal.
	- . A distortion in the shape of the response if the phase response is not a linear function of ω . В.
	- C. The attenuation constant causes phase distortion and the phase constant causes frequency distortion.
	- D. A system introduces magnitude distortion if $|H(\omega)| \neq a$ constant.
	- E. Ideal frequency-selective filters have a distortionless response over one or more frequency bands and zero response elsewhere.

難易度:未設定

A good filter should have a (a) ripple in the passband, (b) attenuation in the stopband and very (c) transition bands.

- what are (a) , (b) , (c) ?
- A. small, high, narrow
- B. small, low, wide
- C. large, high, narrow
- D. small, high, wide
- E. large, low, wide

If $X(z)$ has M finite zeros and N finite poles, then which of the following condition is true?

(a) $|N-M|$ zeros at origin (if $N > M$) (b) $|N-M|$ zeros at origin (if $N < M$) (c) $|N+M|$ zeros at origin (if $N > M$) (d) $|N+M|$ zeros at origin (if $N < M$) (e) $|N-M|$ poles at origin (if $N > M$) (f) $|N-M|$ poles at origin (if $N < M$) (g) $|N+M|$ poles at origin (if $N > M$) (h) $|N+M|$ poles at origin (if $N < M$)

Partial Fraction Expansion $X(z) = \frac{b_0 + b_1 z^{-1} + \dots + b_M z^{-M}}{a_0 + a_1 z^{-1} + \dots + a_N z^{-N}} \rightarrow X(z) = \frac{z^N (b_0 z^M + \dots + b_M)}{z^M (a_0 z^N + \dots + a_N)}$ Hence, it has *M* zeros (roots of $\sum b_k z^{M-k}$), *N* poles (roots of $\sum a_k z^{N-k}$), and (M-N) poles at zero if M>N (or (N-M) zeros at zero if $N>M$).

Given the z-transform pair $x[n] \leftrightarrow X(z)=\dfrac{1}{1-\frac{1}{2}z^{-1}}$ with $ROC:|z|>\dfrac{1}{3}$, determine which z-transform properties can be easily applied to estimate the z-transform of $y[n] = x[n] * x[-n-1]$

(No partial points, you have to select all correct answers.)

(多選題)

- A. Convolution
- **B.** Time shifting
- C. Folding
- D. Differentiation
- \Box E. Linearity
- F. Scaling

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F. Scaling

Solution: Convolution, time shifting and folding: $Y(z) = X(z)X(1/z)z = \frac{-3}{1-3z^{-1}}\frac{1}{1-\frac{1}{2}z^{-1}} = \frac{-3}{1-\frac{10}{2}+z^{-2}}$ ROC: $\frac{1}{3}$ < |z| < $\frac{2}{3}$

11 Given the z-transform pair $x[n] \leftrightarrow X(z) = \frac{1}{1-\frac{1}{2}z^{-1}}$ with $ROC:|z| > \frac{1}{2}$, determine which z-transform properties can be easily applied to estimate the sequence corresponding to $Y(z) = \frac{dX(z)}{dz}$. (No partial points, you have to select all correct answers.) (多選題) $Y(z) = dX(z)/dz$ A. Convolution B. Time shifting Differentiation and time-shifting: $y[n] = -(n-1)x[n-1]$ \Box C. Folding $=-(n-1)\left(\frac{1}{2}\right)^{n-1}u[n-1]$ D. Differentiation E. Linearity

 $12₂$

The one-sided z-transform of $x[n] = \delta[n-k]$ is $\frac{z \wedge (-k)}{k}$, and the one-sided z-transform of $x[n] = \delta[n+k]$ is $\frac{0}{k}$. (忽略空白)

Answering format:

- · real numbers: a or -a
- imaginary numbers: a±bj or -a±bj
- exponential numbers: a^b or a^b (-b) or $(-a)^{b}$ or $(-a)^{b}$ (-b)
- · if a, b are not integers, represent by irreducible fraction, eg., 2/5, 9/4
- no spaces

Let the system is
$$
y[n] = \frac{1}{2}(x[n] + x[n-1])
$$

 \circ D. $\ominus \frac{\omega}{2}$

Difficulty: Not specified

15 The frequency response of the system is:

A
$$
\frac{1}{3}(1 + \cos \omega - j \sin \omega)
$$

\nB. $\frac{1}{3}(1 + \cos \omega + j \sin \omega)$
\nC. $\frac{1}{2}(1 + \cos \omega + j \sin \omega)$
\nD. $\frac{1}{2}(1 + \cos \omega - j \sin \omega)$

Solution:

The system function is:

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

The frequency response is:

$$
H(e^{j\omega}) = \frac{1}{2}(1 + e^{-j\omega}) = \frac{1}{2}(1 + \cos \omega - j\sin \omega)
$$

The magnitude response is:

$$
|H(e^{j\omega})| = \frac{1}{2}\sqrt{(1+\cos\omega)^2 + \sin^2\omega} = |\cos\frac{\omega}{2}|
$$

The phase response is:

$$
\angle H(e^{j\omega}) = \tan^{-1} \frac{-\sin \omega}{1 + \cos \omega} = -\tan^{-1} \tan \frac{\omega}{2} = -\frac{\omega}{2}
$$

For the following input-output pairs determine whether or not there is an LTI system producing y[n] when the input is x[n]. If such a system exists, determine if it is unique or not.

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Consider a second-order IIR filter specification that satisfies the following requirements: (1) the magnitude response is 0 at $\omega_1 = \pi/2$ and $\omega_4 = 3\pi/2$; (2) the maximum magnitude response is 1 at $\omega_{2,3} = \pm 2\pi/3$; and (3) the magnitude response is approximately 1/ V2 at frequencies $\omega_{2,3} \pm 0.05$.

Using the pole-zero placement approach determine locations of two zeros of the required filter:

\n- \n
$$
A_{z_1} = e^{-j\frac{\pi}{2}}
$$
, \n $z_2 = e^{j\frac{3\pi}{2}}$ \n
\n- \n $B_{z_1} = e^{-j\frac{\pi}{2}}$, \n $z_2 = e^{-j\frac{3\pi}{2}}$ \n
\n- \n $C_{z_1} = e^{j\frac{\pi}{2}}$, \n $z_2 = e^{j\frac{3\pi}{2}}$ \n
\n- \n $D_{z_1} = e^{j\frac{\pi}{2}}$, \n $z_2 = e^{\frac{\pi}{2}}$ \n
\n

Difficulty: Not specified

20 Using the pole-zero placement approach determine locations of two poles of the required filter:

A
$$
p_1 = re^{j\frac{\pi}{2}}
$$
, $p_2 = re^{\frac{-j\pi}{2}}$
\nB $p_1 = re^{-j\frac{\pi}{2}}$, $p_2 = re^{j\frac{2\pi}{3}}$
\nC $p_1 = re^{j\frac{\pi}{2}}$, $p_2 = re^{-j\frac{2\pi}{3}}$
\nD $p_1 = re^{j\frac{2\pi}{3}}$, $p_2 = re^{-j\frac{2\pi}{3}}$

Difficulty: Not specified

Solution:
\n
$$
\text{zeros: } z_1 = e^{j\frac{\pi}{2}}, z_2 = e^{j\frac{3\pi}{2}}
$$
\n
$$
\text{poles: } p_1 = re^{j\frac{2\pi}{3}}, p_2 = re^{-j\frac{2\pi}{3}}
$$
\nThe system function is:

\n
$$
H(z) = b_0 \frac{(1 - e^{j\frac{\pi}{2}}z^{-1})(1 - e^{j\frac{3\pi}{2}}z^{-1})}{(1 - re^{j\frac{2\pi}{3}}z^{-1})(1 - re^{-j\frac{2\pi}{3}}z^{-1})}
$$

2 point(s)

$O_{II}175$

A continuous-time LTI system is described by the impulse response $h(t) = 4e^{-0.05t} \cos\left(10\pi t + \frac{\pi}{4}\right)u(t)$. The input to the system is $x(t) = 4 - 3\cos\left(4\pi t + \frac{\pi}{2}\right) + 5\sin(20\pi t)$.

Answering format:

- can be real numbers (integers or fractions of integers): 6 or -6 or 6/4 or -6/4
- can be imaginary numbers: 6j or -6j or (6/4)j or -(6/4)j (remember to use bracket when there are fraction imaginary values)
- no spaces

Determine the system function H(s).

$$
H(s) = \frac{2e^{j a \pi}}{s+b+c\pi} + \frac{2e^{j a \pi}}{s+e+f\pi}
$$

fill the a, b, c, d, e, f of the H(s) shown in above format: a: 1/4, b: 1/20, c: -10, d: -1/4, e: 1/20, f: 10

[Reminder: careful with positive and negative signs eg: 6 or -6; and only use fraction format for float (decimal) numbers, eg: 1/2 not 0.5 (Not case sensitive, Ignoring space)

Solution: $h(t) = 2e^{\int \frac{\pi}{4}} e^{-0.005t} e^{j10\pi t} u(t) + 2e^{-\int \frac{\pi}{4}} e^{-0.005t} e^{-j10\pi t} u(t)$ The system function is: $H(s) = \frac{2e^{j\frac{\pi}{4}}}{s+0.05-10\pi} + \frac{2e^{-j\frac{\pi}{4}}}{s+0.05+10\pi}$

For a distortionless LTI system: 24

- A. group delay = phase delay
- B. group delay > phase delay
- \circ C. group delay < phase delay
- \circ D. group delay = phase delay/2

Difficulty: Not specified

25

 3 point(s)

Group delay concept

page 36-38

from June01-2022.pdf

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The response of a stable LTI system to a complex exponential sequence is a complex exponential sequence with the same (a) , only the (a) $_$ and $_$ (c) $_$ are changed by the system. what are (a),(b),(c)?

- A. phase, frequency, amplitude
- B. frequency, amplitude, phase \circledcirc
- C. frequency, amplitude, group delay
- © D. amplitude, phase, frequency,

Difficulty: Not specified