

Table No.:	
------------	--

NAME :
SID :
SECTION :

Q1	
Q2	
Q3	
Q4	
Total	



**COLLEGE OF ENGINEERING
PUTRAJAYA CAMPUS
TEST 2
SEMESTER 1, 2021/2022**

PROGRAMME : Bachelor of Electrical & Electronics Engineering (Honours) /
Bachelor of Electrical Power Engineering (Honours)
SUBJECT CODE : EEEB363/EEEB3024
SUBJECT : Digital Signal Processing
DATE : 29th December 2021
DURATION : 2.5 hours

INSTRUCTIONS TO CANDIDATES:

1. This paper contains **FOUR (4)** questions.
2. **Answer all questions.**

DO NOT OPEN THE QUESTION PAPER UNTIL YOU ARE TOLD TO DO SO

**THIS QUESTION PAPER CONSISTS OF FIVE (5) PRINTED PAGES
INCLUDING THIS COVER PAGE**

QUESTION 1 [20 MARKS]

(a) A causal finite impulse response filter is given as $h_1[n] = \{ 1, 3, 3, 1 \}; 0 \leq n \leq 3$.

- (i) Write $h_1[n]$ as a function of delayed unit impulses, $\delta[n]$. Then determine its z-transform, $H_1(z)$.

[3 marks]

- (ii) Sketch the poles of $H_1(z)$ on a z-plane.

[3 marks]

- (iii) Describe the stability and causality of the finite impulse response filter above based on its region of convergence (ROC).

[2 marks]

(b) A digital filter transfer function is given by

$$H_2(z) = \frac{z^2 - 2z - 3}{z^2 + 0.8z + 0.15}$$

- (i) Sketch the poles of $H_2(z)$ on another z-plane.

[3 marks]

- (ii) Determine the various possible ROC for $H_2(z)$.

[3 marks]

(c) A digital filter transfer function is given by

$$H_3(z) = \frac{1}{1 - 0.8 e^{-j\frac{\pi}{2}} z^{-1}} + \frac{1}{1 - 0.8 e^{j\frac{\pi}{2}} z^{-1}}$$

- (i) Sketch the poles of $H_3(z)$ on another z-plane.

[3 marks]

- (ii) If the system $H_3(z)$ is stable and causal, determine the impulse response $h_3[n]$.

[3 marks]

QUESTION 2 [30 MARKS]

- (a) A linear time-invariant (LTI) discrete-time system is described in **Figure 1**.

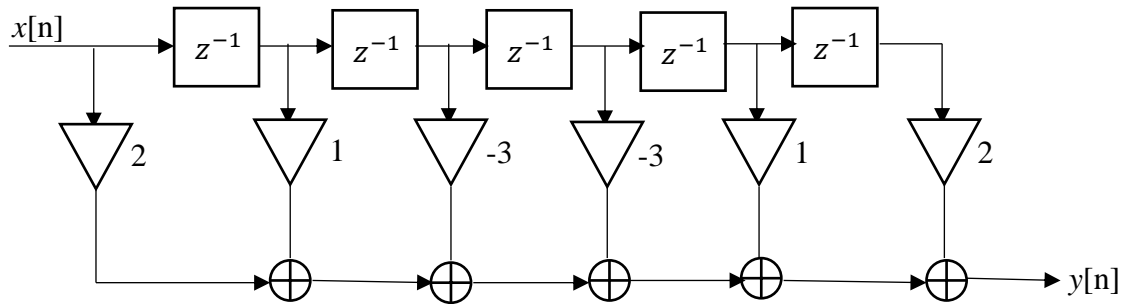


Figure 1: Linear Time Invariant (LTI) discrete-time system

- (i) Analyze the block diagram in **Figure 1** and develop the relation between output, $y[n]$ and input, $x[n]$.
[3 marks]
 - (ii) Determine the impulse response, $h[n]$, of above LTI discrete-time system.
[5 marks]
 - (iii) Determine **AND** explain the linearity and causality condition of $h[n]$.
[4 marks]
 - (iv) Determine the group delay value of $h[n]$.
[3 marks]
- (b) A typical transmission channel is characterized by a causal transfer function, $H(z)$.

$$H(z) = \frac{(0.5z - 1)(4z - 2)(2z + 3)}{(z - 0.6)(z + 0.4)(z + 0.2)}$$

- (i) Determine all poles **AND** zeros value of $H(z)$.
[3 marks]
- (ii) Classify $H(z)$ as a minimum, maximum or mixed phase system. In your own words, discuss on the choice of phase class.
[4 marks]
- (iii) Represent $H(z)$ as a minimum phase function, $H_{min}(z)$ in cascade with linear phase function, $H_{lin}(z)$.
[8 marks]

QUESTION 3 [25 MARKS]

Consider the cascade of two causal first order and second order linear time-invariant (LTI) discrete time system shown in **Figure 2**.

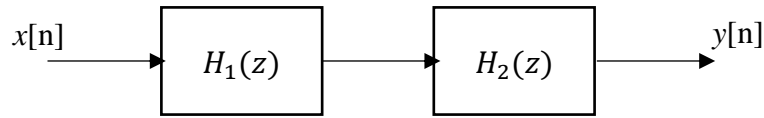


Figure 2: First order and second order LTI-DT system

where,

$$H_1(z) = \frac{2 - 0.3z^{-1}}{1 + 0.5z^{-1}} \text{ and } H_2(z) = \frac{(1 - 3z^{-1})(1 + z^{-1})}{(1 + 0.3z^{-1})(1 - 0.2z^{-1})}$$

- (a) Determine the transfer function of the overall cascade system, $H(z) = H_1(z) \times H_2(z)$.

[3 marks]

- (b) Develop the realization of the overall system, $H(z)$ in Direct Form I **AND** Direct Form II.

[12 marks]

- (c) In your own words, describe the differences between Direct Form I and Direct Form II.

[4 marks]

- (d) Develop the cascade form realization of $H(z)$ with the least number of coefficients in the structure.

[6 marks]

QUESTION 4 [25 MARKS]

- (a) When selecting digital filters for an application, compare between finite impulse response (FIR) and infinite impulse response (IIR) filters as below:

- (i) Discuss the differences between FIR and IIR based on their phase response and the number of coefficients.

[2 marks]

- (ii) What needs to be done in the design steps to narrow the transition bandwidth in FIR and IIR filters respectively? Please explain.

[2 marks]

- (b) Somebody hacked your Discord server and now all voice chats have some audio noise at a specific frequency. You recorded the noise and managed to identify the noise frequencies by analyzing its frequency spectrum. Following your findings, a digital filter is proposed to attenuate the noise, with a sampling frequency of 1000Hz and the following specifications:

$$0.90 \leq |H(e^{j\omega})|^2 \leq 1.00 ; |f| \leq 300\text{Hz}$$

$$|H(e^{j\omega})|^2 \leq 0.10 ; |f| \geq 400\text{Hz}$$

- (i) According to the specifications mentioned above, sketch the expected graph / tolerance diagram of its squared magnitude response $|H(e^{j\omega})|^2$ versus normalized angular frequency ω . Carefully label the gains, passband edge frequency, stopband edge frequency, passband and stopband frequency ranges.

[6 marks]

- (ii) Design a Butterworth infinite impulse response (IIR) filter using the above specifications.

[15 marks]