

Roll No.

8616

Printed Pages : 4

BT-6 / M12

DIGITAL SIGNAL PROCESSING

Paper-ECE-306-E

Time allowed : 3 hours

Maximum marks : 100

Unit-I

1. (a) Determine the causal signal $x(n]$ if its z-transform $X(z)$ is given by :

$$X(z) = \frac{1}{4} \frac{(1 + 6z^{-1} + z^{-2})}{(1 - 2z^{-1} + 2z^{-2}) \left(1 - \frac{1}{2}z^{-1}\right)} \quad 5$$

- (b) If $X(z)$ is the z-transform of $x(n]$, show that if

$$x_k(n) = \begin{cases} x\left(\frac{n}{k}\right), & \text{if } \frac{n}{k} \text{ integer} \\ 0, & \text{otherwise} \end{cases}$$

then $X_k(Z) = X(Z^k)$ 8

- (c) Determine the convolution of following pairs of signals by means of z-transform

$$x_1(n) = nu(n)$$

$$x_2(n) = (2^n)u(n-1) \quad 7$$

8616-0-8-6950

(P.T.O)

(2)

2. (a) Compute the unit step response of the system with the impulse response 8

$$h(n) = \begin{cases} 3^n, & n < 0 \\ \left(\frac{2}{5}\right)^n, & n \geq 0 \end{cases}$$

- (b) Determine if the following FIR system is minimum phase or not : 6

$$h(n) = \{10, 9, -7, -8, 0, 5, 3\}$$

- (c) Compute the quantity 6

$$\sum_{n=0}^{(N-1)} x_1(n)x_2(n) \text{ if}$$

$$x_1(n) = \cos\left(\frac{2\pi}{N}n\right), x_2(n) = \sin\left(\frac{2\pi}{N}n\right), 0 \leq n \leq (N-1)$$

Unit-II

3. (a) Determine a direct form realization of following linear phase filter 5

$$h(n) = \left\{ \underset{\uparrow}{1}, 2, 3, 4, 3, 2, 1 \right\}$$

- (b) Determine the state-space model for the system described by

$$y(n) = y(n-1) + 0.11y(n-2) + x(n)$$

and sketch the type 1 and type 2 state-space realizations. 5+5+5

8616

(3)

4. (a) Consider the system described by the difference equation

$$y(n) = a y(n-1) - a x(n) + x(n-1)$$

- (i) Show that it is all-pass. 5
 (ii) Obtain the direct form II realization of the system. http://www.kuonline.in 5

- (b) Consider a causal IIR system with the system function

$$H(z) = \frac{1 + 2z^{-1} + 3z^{-2} + 2z^{-3}}{1 + 0.9z^{-1} - 0.8z^{-2} + 0.5z^{-3}}$$

Determine the equivalent lattice-ladder structure. 10

Unit-III

5. (a) Design an FIR linear-phase, digital filter approximating the ideal frequency response

$$H_d(\omega) = \begin{cases} 1 & \text{for } |\omega| \leq \frac{\pi}{6} \\ 0 & \text{for } \frac{\pi}{6} < |\omega| \leq \pi \end{cases}$$

- (i) Determine the coefficients of a 25-tap filter based on window method with a rectangular window. 5
 (ii) Repeat part (i) using Hamming window. 3

- (b) Explain the Gibbs phenomenon with example. 10

8616

[P.T.O.]

(4)

6. (a) Write a short note on "Alternation theorem". 10
 (b) Determine the unit sample response $\{h(n)\}$ of a linear-phase FIR filter of length $M = 9$ for which the frequency response at $\omega = 0$ and $\omega = \frac{\pi}{2}$ is specified as $H_r(0) = 1$, $H_r\left(\frac{\pi}{2}\right) = \frac{1}{2}$. 10

Unit-IV

7. Explain the design of digital filters based on least square's methods. 20
 8. A digital low-pass filter is required to meet the following specifications:

Passband Ripple : ≤ 1 dB

Passband Edge : 4 KHz

Stopband attenuation : ≥ 40 dB

Stopband edge : 6 KHz

Sample Rate : 24 KHz ; Assume $t = 1$

The filter is to be designed by performing a bilinear transformation on an analog system function. Determine what order Butterworth, Chebyshev and Elliptic analog designs must be used to meet the specifications in the digital implementation. 6+6+8

8616