

Total No. of Questions :10]

SEAT No. :

**P2885**

[Total No. of Pages :3

[4958] - 1076

**T.E. (Instrumentation & Control)**  
**DIGITAL SIGNAL PROCESSING**  
**(Semester - II) (2012 Course)**

*Time : 2½ Hours]*

*[Max. Marks :70*

*Instructions to the candidates:*

- 1) *Answer Q1 or Q2, Q3 or Q4, Q5 or Q6, Q7 or Q8, Q9 or Q10.*
- 2) *Neat diagrams must be drawn wherever necessary.*
- 3) *Figures to the right side indicate full marks.*
- 4) *Assume Suitable data if necessary.*

**Q1) a)** Explain in detail classification of system. **[4]**

b) If the impulse response of the system is **[6]**

$$h(n) = [(0.5)^n + n(0.2)^n] u(n)$$

i) Find the Transfer Function

ii) Obtain the differential equation of the system.

OR

**Q2) a)** Find inverse z- transform of **[5]**

$$X(z) = \frac{z(z-0.5)}{(z-0.8)(z-1)}$$

b) Find DFT of a sequence for N = 4 **[5]**

$$X(n) = \begin{cases} 0.5 & 0 \leq n \leq 2 \\ 0 & \text{otherwise} \end{cases}$$

**P.T.O.**

**Q3) a)** Find impulse and step response for the following system [6]

$$X(n) = y(n) - 3/4 y(n-1) + 1/8 y(n-2).$$

b) Compute 4 - point DFT of the sequence [4]

$$X(n) = \{3, 1, 5\}.$$

OR

**Q4) a)** Determine the Circular convolution of the given two sequence. [6]

$$X(n) = \{1, 2, 3, 2\} \quad h(n) = \{1, 2, 1, 2\}.$$

b) Explain any two properties of Discrete Time Fourier Transform (DTFT) with proof. [4]

**Q5) a)** Obtain the 8-point DFT of a given sequence using DIT FFT Algorithm [12]

$$X(n) = \{1, 1, 2, 1\}.$$

b) Explain the concept of bit reversal technique in FFT Algorithm. [4]

OR

**Q6) a)** Obtain the 4-point DFT of a given sequence using DIT & DIF FFT Algorithm [12]

$$X(n) = \{4, 3, 2, 1\}.$$

b) Sketch Signal flow graph of 8-point Decimal in Frequency (DIF) FFT Algorithm. [4]

- Q7) a)** Explain various window functions used in FIR filter design. [6]
- b) Design a low pass filter with 11 coefficients for following specification  
 Pass band frequency edge = 250 Hz, Sampling frequency = 1000 Hz  
 Use Rectangular and Hamming Window for Design. [12]

OR

- Q8) a)** Design a linear phase FIR filter using Hamming window for desired frequency response [12]

$$H_d(\omega) = e^{-j(N-1)\omega/2}, \quad 0 < |\omega| < \pi/4$$

$$0, \quad \pi/4 < |\omega| < \pi \quad \text{for } N = 7$$

- b) Differentiate between FIR & IIR filter. [6]
- Q9) a)** Find the order of analog low pass Butterworth filter for the given specification [6]

$$\alpha_p = 1 \text{ dB}, \alpha_s = 30 \text{ dB}, \Omega_p = 200 \text{ rad/sec}, \Omega_s = 600 \text{ rad/sec.}$$

- b) Design a Chebyshev filter with a max pass band attenuation of 2.5 dB at  $\Omega_p = 20 \text{ rad/sec}$  & stop band attenuation of 30 dB at  $\Omega_s = 50 \text{ rad/sec}$ . [10]

OR

- Q10) a)** Design a Butterworth filter using Impulse invariance method for following specification. [10]

$$0.9 \leq |H(e^{j\omega})| \leq 1 \quad 0 \leq \omega \leq \pi/2$$

$$|H(e^{j\omega})| \leq 0.2 \quad 3\pi/4 \leq \omega \leq \pi$$

- b) Compare the features of digital Butterworth and Chebyshev type-1 filter in terms of [6]
- i) Filter order
- ii) Transition width

