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**EX - 703**

**B.E. VII Semester**

Examination, December 2012

**Digital Signal Processing**

*Time : Three Hours*

*Maximum Marks : 100*

*Minimum Pass Marks :35*

*Note:* Attempt *One* question from each unit  
All questions carry equal marks.

**UNIT I**

- I (a) Examine the following systems with respect to the properties - linearity, Time variance, causality and stability.

(i)  $y(n) = x(-n+2)$

(ii)  $y(n) = x(2n)$

(iii)  $y(n) = \sum_{k=-\infty}^n x(k)$

- (b) Determine the Fourier transform of the signal

$$x(n) = a^{|n|}, -1 < a < 1$$

OR

2. (a) State and prove the following properties of DTFT

(i) Multiplication of two sequence

(ii) Differentiation in the frequency domain

- (b) Discuss discrete time processing of continuous time signals

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**UNIT II**

3. (a) Determine the step response of the system

$$y(n) = \alpha y(n-1) + x(n) - 1 < \alpha < 1$$

when the initial condition is  $y(-1) = 1$

- (b) Determine the transient and steady state responses of the system characterized by the difference equation.

$$y(n) = 0.5 y(n-1) + x(n)$$

when the input signal is  $x(n) = 10 \cos(\pi n/4) u(n)$ . The system is initially at rest

OR

4. A LTI system is characterised by the system function

$$H(z) = \frac{3 - 4z^{-1}}{1 - 3.5z^{-1} + 1.5z^{-2}}$$

Specify the ROC of  $H(z)$  and determine  $h(n)$  for the following condition

- (i) The system is stable
- (ii) The system is causal
- (iii) The system is anti causal

**UNIT III**

5. (a) Perform the circular convolution of the following sequences.

$$x_1(n) = \{2, 1, 2, 1\}$$

↑

$$x_2(n) = \{1, 2, 3, 4\}$$

↑

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- (b) Derive the signal flow graph for the  $N=16$  point, radix-4, decimation in time FFT algorithm

OR

6. Compute the eight point DFT of the sequence

$$x(n) = \begin{cases} 1, & 0 \leq n \leq 7 \\ 0, & \text{otherwise} \end{cases}$$

by using the decimation in frequency FFT algorithm

**UNIT IV**

7. (a) Explain the bilinear transformation method for designing IIR filter.

- (b) Convert the analog filter with system function

$$H_a(s) = \frac{s + 0.1}{(s + 0.1)^2 + 9}$$

$$(s + 0.1)^2 + 9$$

into a digital IIR filter by means of the impulse invariance method.

OR

8. (a) Discuss Butterworth analog filter approximation for IIR digital filter.

- (b) Explain canonic realization of IIR digital filter

**UNIT V**

9. Discuss the designing of FIR filter using Kaiser window.

OR

10. Discuss the designing of FIR filter using rectangular window.

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