

**R13**

Code No: 126EK

JAWAHARLAL NEHRU TECHNOLOGICAL UNIVERSITY HYDERABAD

B. Tech III Year II Semester Examinations, October/November - 2016

DIGITAL SIGNAL PROCESSING

(Common to ECE, EIE)

Time: 3 hours

Max. Marks: 75

**Note:** This question paper contains two parts A and B. Part A is compulsory which carries 25 marks. Answer all questions in Part A. Part B consists of 5 Units. Answer any one full question from each unit. Each question carries 10 marks and may have a, b, c as sub questions.

**PART - A****(25 Marks)**

- 1.a) Define decimation and interpolation. [2]
- b) State linear and static system with example. [3]
- c) If  $x(n) = \cos\left(\frac{\pi}{3}n\right)$ , find spectra of the signal? [2]
- d) How many multiplications and additions are required to compute N-point DFT using radix-2 FFT? [3]
- e) State the properties of IIR filter. [2]
- f) State the methods used to prevent overflow. [3]
- g) Give the equations for Hamming window and Blackman window [2]
- h) What is dead band of a filter? [3]
- i) Give the steps in multistage sampling rate converter design. [2]
- j) Write any four applications of Multi-rate signal processing. [3]

**PART - B****(50 Marks)**

- 2.a) If  $y(n) = 12x(n-1) + 11x(n-2)$ , Find whether given system is time variant or not?
- b) Find the frequency response of 1<sup>st</sup> order system  $y(n) = x(n) + ay(n-1)$ . [5+5]

**OR**

- 3.a)  $y(n) - 3y(n-1) - 4y(n-2) = 0$  determine zero-input response of the system; Given  $y(-2) = 0$  and  $y(-1) = 5$ .
- b) Obtain the cascade and parallel form realization for the system.  
 $y(n) = -0.1y(n-1) + 0.2y(n-2) + 3x(n) + 3.6x(n-1) + 0.6x(n-2)$ . [5+5]

4. Explain all properties of DTFT. [10]

**OR**

- 5.a) Explain Radix-2 Decimation-in-time FFT algorithm with example.
- b) Find the DFT of a sequence  $x(n) = \{1, 1, 0, 0\}$  and find the IDFT of  $Y(k) = \{1, 0, 1, 0\}$ . [5+5]

6. Design a Butterworth high pass filter satisfying:  
 $f_p = 0.32\text{Hz}$ ;  $\alpha_p = 0.5\text{ dB}$   
 $f_s = 0.16\text{Hz}$ ;  $\alpha_s = 30\text{ dB}$ ;  $F = 1\text{ Hz}$  [10]

**OR**

7. Discuss the steps in design of IIR filter using bilinear transformation for any one type of filter. [10]

8. Realize the system function  $\frac{2}{3}z + 1 + \frac{2}{3}z^{-1}$  by linear phase FIR structure. [10]

OR

9. Explain the designing of FIR filters using windows. [10]

10. With respect to finite word length effects in digital filters, explain over flow limit cycle oscillation with example. [10]

OR

11.a) Explain the applications of Multi rate signal processing.

b) What is meant by signal scaling? Explain. [5+5]

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