

Total No. of Questions : 12]

SEAT No. :

P2450

[5153]-84

[Total No. of Pages : 2

T.E.(Computer)

DIGITAL SIGNAL PROCESSING

(2008 Pattern) (Semester-I)

Time : 3 Hours]

[Max. Marks :100

Instructions to the candidates:

- 1) *Answers to the two sections should be written in separate answer books.*
- 2) *Answer any three questions from each section.*
- 3) *Neat diagrams must be drawn wherever necessary.*
- 4) *Figures to the right side indicate full marks.*
- 5) *Use of Calculator is allowed.*
- 6) *Assume suitable data, if necessary.*

SECTION-I

Q1) Explain different classification of signals with example. **[16]**

OR

Q2) a) Define energy and power signal. Find whether the following signal is an energy signal or power signal $x(n) = n$ $n > 0$
 $= 0$ $n < 0$ **[7]**

b) Test Linearity, Causality, and time invariance systems of $y(n)=x(n) u(n)$. **[9]**

Q3) a) Obtain DTFT of different standard signals. **[12]**

b) Explain clearly: Circular convolution. **[6]**

OR

Q4) a) Compare DFT with DTFT. **[4]**

b) Why the result of circular and linear convolution is not same? **[4]**

c) Using DFT and IDFT obtain circular convolution of $x(n) = \{2, 0, 0, 1\}$ and $h(n) = \{4, 3, 2, 1\}$. **[10]**

Q5) Derive and explain Radix-2 Decimation in time (DIT) FFT algorithm for computing N-point DFT. **[16]**

OR

Q6) a) State time shifting property of Z-transform and calculate $x(n) = \delta(n+2)$. **[4]**

b) Explain different properties of twiddle factor. **[9]**

c) Define ROC and state significance of ROC. **[3]**

P.T.O.

SECTION-II

Q7) a) Explain the method of simple geometric interpretation to obtain the frequency response of DT system. [10]

b) Determine impulse response of a system:

$$y(n)=x(n)-x(n-1)-3y(n-1)-2y(n-2) \quad [8]$$

OR

Q8) a) Define system function $H(z)$. How it is obtained from the general difference equation? [6]

b) what is pole zero plot? State condition for causality and stability in terms of ZT. [6]

c) Determine $H(z)$ and draw a pole zero plot for a system [6]

$$x(n)+x(n-1)=y(n)+\frac{3}{4}y(n-1)+\frac{1}{8}y(n-2)$$

Q9) a) Explain Gibb's phenomenon associated with FIR filter design. What are the desirable features of window function to improve the frequency response? [10]

b) The transfer function of analog filter is: $H(s)=\frac{3}{(s+2)(s+3)}$ with

$T_s=0.1$ sec. Design the digital IIR filter using BLT method. [6]

OR

Q10) a) To design the digital IIR filter, analog IIR filter is designed first, why? What are the different methods to design IIR filters? Explain any one in brief. [10]

b) What are the advantages and disadvantages of FIR filter. [6]

Q11) a) Obtain linear phase FIR filter of $H(z)=\left(1+\frac{z^{-1}}{4}+\frac{z^{-2}}{4}+z^{-3}\right)$. [6]

b) Compare DSP processor and general purpose processors. [6]

c) What are the advantages of representing the digital filter in block diagram form? [4]

Q12) a) Draw Direct Form-I IIR filter structure for: $H(z)=3+\frac{4z}{(z-0.5)}-\frac{2}{(z-0.25)}$ [8]

b) Explain the application of DSP in speech processing. [8]

