

T.E. (Electronics)

DISCRETE TIME SIGNAL PROCESSING**(2008 Pattern) (Semester - II) (304211) (Theory) (BOS)***Time : 3 Hours]**[Max. Marks :100**Instructions to the candidates:*

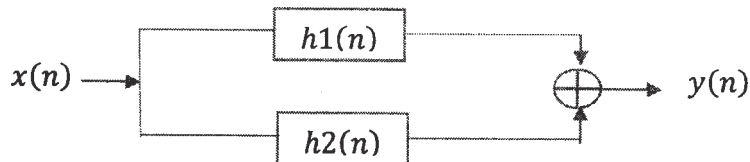
- 1) Attempt Q.1 or Q.2, Q.3 or Q.4, Q.5 or Q.6, Q.7 or Q.8, Q.9 or Q.10, Q.11 or Q.12.
- 2) Answers to the two sections should be written in separate answer books.
- 3) Neat diagrams must be drawn wherever necessary.
- 4) Figures to the right side indicate full marks.
- 5) Use of electronic pocket calculator is allowed.
- 6) Assume suitable data, if necessary.

SECTION - I

- Q1)** a) Two signals $x_1(t) = \cos(20\pi t)$ and $x_2(t) = \cos(100\pi t)$ with sampling frequency 40 Hz. Obtain the associated discrete time signals & comment on result. [8]
- b) Explain Stability & causality of LTI system in terms of its impulse Response. [8]

OR

- Q2)** a) Find out the particular solution for [6]
 $y(n) + 3y(n-1) = x(n)$ Assume $x(n) = u(n)$
- b) If LTI system is described as [6]



If $x(n) = \{1, 2, 4\}$ & $h_1(n) = \{1, 2, 0, 1\}$, $h_2(n) = \{2, 1, 1, 2\}$

- c) Explain the basic elements of DSP with the help of block diagram. [4]

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- Q3)** a) Explain the Relation between DTFT & DFT. [4]
 b) Determine 2-point & 4-point DFT of $x(n) = u(n) - u(n-2)$ [6]
 c) Explain overlap save method in detail. [8]

OR

- Q4)** a) Compute the DFT $x(n) = \cos(n\pi/2)$ for $N=4$ using DIF-FFT Algorithm. [8]
 b) Obtain the Circular convolution of $x_1(n) = \{1, 2, 3, 1\}$ & $x_2(n) = \{4, 3, 2, 2\}$ By using matrix multiplication method. [4]
 c) Explain cyclic Property of Twiddle Factor for $N=4$. [6]

- Q5)** a) If $x(z) = 2 + 3z^{-1} + 4z^{-2}$ find the initial & final values of the cross ponding sequence $x(n)$. [6]
 b) Given that [10]

$$H(z) = \frac{-4 + 8z^{-1}}{1 + 6z^{-1} + 8z^{-2}} \text{ is casual system find}$$

- i) Transfer function Representation
 ii) Difference Equation Representation
 iii) Impulse Response Representation.

OR

- Q6)** a) Determine the Z-transform of the following sequence using Properties only [10]
 i) $u(n-4)$
 ii) $\delta(n-5)$
 iii) $e^{jn\pi/4} u(n)$
 iv) $(1/3)^n u(-n)$
 v) $3^n u(n-2)$
 b) Find the convolution of $x(n) = \{1, 2, 3, 1, -1, 1\}$ & $h(n) = \{1, 1, 1\}$ using Z-Transform. [6]

SECTION - II

- Q7)** a) Define the terms Related to FIR filter [6]
- i) Phase Delay
 - ii) Group Delay
 - iii) Symmetric filter
 - iv) Antisymmetric filter
 - v) Linear phase filter
 - vi) Order of filter
- b) Design an FIR Digital filter to approximate an LPF with Pass band gain unity, cut of frequency of 850 HZ and working at a sampling frequency of 5000HZ. The length of impulse response should be 5. use Hamming Window. [12]

OR

- Q8)** a) For the analog filter $H(S) = \frac{1}{(S+1)(S+2)}$. Convert above analog filter into Digital Filter by using [10]
- i) Impulse invariance method.
 - ii) BLT method
- b) Prove along with mathematical expression “*Stable Analog filter is converted into stable Digital Filter using BLT method*”. [8]
- Q9)** a) Explain the Application of Multirate sampling in Data Acquisition System along with Block diagram. [8]
- b) Explain the Role of Antialiasing Filter in Decimator & Role of Anti-imaging filter in interpolator in Detail. [8]

OR

Q10)a) Design two stage decimator with sampling rate to be reduced from 10KHz to 500 Hz. Passband edge of 150 Hz, stopband edge of 180 Hz, Passband ripple 0.002 & stopband ripple 0.001 & $D_1 = 10$, $D_2 = 2$. [10]

b) Explain sampling rate conversion By I/D factor. [6]

Q11)a) Explain the architecture of TMS320C28XX in detail. [8]

b) Differentiate between General Purpose Microprocessor & DSP processor. [8]

OR

Q12)a) Explain Related to DSP processor. [8]

i) DAG

ii) Barrel Shifter

b) Explain any four addressing Modes of for DSP processor in detail. [8]

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