

**5E5107**  
**B. Tech. V Sem. (Main/Back) Exam., Nov.-Dec.-2016**  
**Computer Science & Engineering**  
**5CS6.2A Digital Signal Processing**  
**Common With CS, IT**

Time: 3 Hours

Maximum Marks: 80  
 Min. Passing Marks Main: 26  
 Min. Passing Marks Back: 24

Instructions to Candidates:

Attempt any **five questions**, selecting **one question** from **each unit**. All questions carry **equal marks**. Schematic diagrams must be shown wherever necessary. Any data you feel missing suitably be assumed and stated clearly.

Units of quantities used/calculated must be stated clearly.

Use of following supporting material is permitted during examination.  
 (Mentioned in form No. 205)

1. NIL2. NIL**UNIT - I**

- Q.1 (a) Check whether the following systems are linear - [8]
- (i)  $F[x(n)] = ax(n) + b$
- (ii)  $F[x(n)] = e^{x(n)}$
- (b) Check whether the following systems are linear & time invariant - [8]
- (i)  $F[x(n)] = n[x(n)]^2$
- (ii)  $F[x(n)] = a[x(n)]^2 + 3(x(n))$

**OR**

- Q.1 (a) List out the properties of LTI system. Explain each with the help of an example. [8]

(b) A second order discrete time system is characterized by the difference equation. [8]

$$y(n] - 0.1y(n - 1) - 0.02y(n - 2) = 2x(n) - x(n - 1)$$

Determine  $y(n)$  for  $n \geq 0$  when  $x(n) = 4^n$  &

The initial conditions are  $y(-1) = -10$  &

$$y(-2) = 5$$

### UNIT - II

Q.2 Find out the linear convolution of an FIR filter with impulse response  $h(n) = \{1, 2, 4\}$  to the input sequence  $x(n) = \{1, 2\}$

(Use process of circular convolution with padding of Zeros.) [16]

OR

Q.2 (a) Define ROC for Z transform. Write down its properties for Z transform. [6]

(b) Determine the inverse Z transform of the following  $x(z)$  by the partial fraction expansion method. [10]

$$\left[ x(z) = \frac{z + 2}{2z^2 - 7z + 3} \right]$$

If the ROCs are

(a)  $|z| > 3$

(b)  $|z| < \frac{1}{2}$

(c)  $\frac{1}{2} < |z| < 3$

### UNIT - III

Q.3 Define sampling theorem. Explain Interpolation technique for the reconstruction of a signal from its samples. [16]

OR

Q.3 (a) What do you mean by aliasing. [6]

(b) A Continuous signal is given as  $x_e(t) = \sin(2\pi(100)t)$

This signal was sampled with sampling period  $T = 1/400$  second to obtain discrete time signal  $x(n)$ . Determine the resulting signal  $x(n)$ . [10]

## UNIT - IV

Q.4 (a) Find the DFT of the sequence:

$$x(n) = \begin{cases} 1; & 2 \leq n \leq 6 \\ 0; & n = 0, 1, 7, 8, 9 \end{cases}$$

[10]

Given  $N = 10$

(b) Explain Parseval's theorem & circular shift property for DFT.

[6]

OR

Q.4 (a) Explain Decimation in frequency FFT (DIF-FFT) algorithm; with diagram. [8]

(b) Determine the four point DFT for  $x(n) = \{0, 1, 2, 3\}$  using DIT - FFT algorithm. [8]

## UNIT - V

Q.5 Write short note on -

[4×4=16]

- (a) Structures of FIR filter
- (b) Cascade & parallel structure
- (c) Hamming window
- (d) Chebyshev filter

OR

Q.5 The following specifications are given to design a band pass filter -

[16]

Stop band frequencies =  $0.1\pi$  rad/sec,

$0.7\pi$  rad/sec.

Pass band lower cutoff frequency =  $0.25\pi$  rad/sec

Pass band upper cutoff frequency =  $0.35\pi$  rad/sec

Pass band attenuation = - 3dB.

Stop band attenuation = - 15dB

Sampling rate = 1 sample/sec.

Design a Band pass filter using bilinear transformation.