

Roll No.

Total No. of Questions : 09]

[Total No. of Pages : 02

Paper ID [A0321]

(Please fill this Paper ID in OMR Sheet)

B.Tech. (Sem. - 6th)

DIGITAL SIGNAL PROCESSING (EC - 308)

Time : 03 Hours

Maximum Marks : 60

Instruction to Candidates:

- 1) Section - A is **Compulsory**.
- 2) Attempt any **Four** questions from Section - B.
- 3) Attempt any **Two** questions from Section - C.

Section - A

Q1)

(10 × 2 = 20)

- a) Write any disadvantage of digital over analog signal processing.
- b) Differentiate time variant from time invariant system?
- c) Define sampling theorem.
- d) Define a Causal system.
- e) Differentiate stable from a non-stable system.
- f) Write application of FFT algorithm?
- g) What is zero padding in DFT?
- h) What is linear convolution.
- i) Write any two basic features of IIR filters.
- j) Write any two applications of Z-Transforms in signal processing.

Section - B

(4 × 5 = 20)

Q2) Show that $h(n)$ is equal to the convolution of the following signals

$$h_1(n) = \delta(n) + \delta(n-1)$$

$$h_2(n) = (1/2)^n u(n).$$

Q3) What is the physical significance of ROC in Z transform.

- Q4) Find out the Z-transform for the following discrete time sequence
 $x(n) = kn^2, n \geq 0$.
- Q5) Discuss FFT algorithm using decimation in frequency technique.
- Q6) Discuss various properties of DFT.

Section - C

(2 × 10 = 20)

- Q7) Discuss signal flow graph representation and lattice form structures for FIR systems.
- Q8) Discuss various steps for the design of linear phase IIR filters by impulse invariance technique.
- Q9) Discuss basic architecture of ADSP series of digital signal processors.

