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Total No. of Pages : 02

Total No. of Questions : 09

B.Tech.(Electronics &amp; Computer Engg.) (2011 Onwards) (Sem.-6)

**DIGITAL SIGNAL PROCESSING**

Subject Code : BTEC-502

Paper ID : [A2347]

Time : 3 Hrs.

Max. Marks : 60

**INSTRUCTIONS TO CANDIDATES :**

1. SECTION-A is COMPULSORY consisting of TEN questions carrying TWO marks each.
2. SECTION-B contains FIVE questions carrying FIVE marks each and students have to attempt any FOUR questions.
3. SECTION-C contains THREE questions carrying TEN marks each and students have to attempt any TWO questions.

**SECTION-A****1 Answer briefly :**

- A What do you mean by LTI system? Explain.
- B What do you mean by cross correlation? Explain.
- C Define DFT and List its computational requirements.
- D Explain time shifting property of Z-transform.
- E What do you mean by pass band ripple? Explain.
- F Write down the various advantages of digital filters.
- G List the properties of Hamming and Hanning windows used for FIR filter design.
- H Write down the various advantages of DSP processor.
- I What do you mean by limit cycles? Explain.
- J Explain the property of linear phase in FIR filters.

**SECTION-B**

- 2 What is digital signal processing? Discuss its different benefits and applications in detail.
- 3 Find the inverse Z-transform of the  $X(z) = \frac{z}{(z - 0.75)(z - 0.5)^2}$
- 4 Calculate the convolution of  $x(n) = \underset{\uparrow}{[3, 4, 7, 6, 5]}$  and  $h(n) = \underset{\uparrow}{[1, 5, 7, 8, 2]}$
- 5 Discuss the design of an IIR filter by impulse invariant method by considering suitable example.
- 6 Determine the direct form-I and direct form -II structure for the systems described by the system function

$$H(z) = \frac{1 - 0.8z^{-1} + 0.15z^{-2}}{1 + 0.1z^{-1} - 0.7z^{-2}}$$

**SECTION-C**

- 7 Compute the 8-point DFT of  $x(n) = 2n + 1$  ;  $0 \leq n \leq 7$  using Decimation in time FFT algorithm.
- 8 Obtain the coefficients of a FIR filter to meet the specifications given below using the window method.

Stopband attenuation > 40 dB

Transition width 0.5 kHz

Sampling frequency 8 kHz

Ideal cutoff frequency 1.5 kHz

- 9 Explain in detail the architecture of ADSP processor with the help of neat sketch.