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B.Tech.(Electronics & Electrical) (2011 Onwards) B.Tech. (Electrical & Electronics) (2013 Batch) (Sem.-6) DIGITAL SIGNAL PROCESSING Subject Code : BTEEE-601 M.Code : 71130

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. Write briefly :
 - a) What are the advantages of digital over analog signal processing?
 - b) Find the even and odd parts of the x(n) = u(n).
 - c) What is the z-transform of the finite duration signal?

$$X(n) = \{2,4,5,7,0,1\}$$
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- d) Why the result of circular and linear convolution is not same?
- e) What are the advantages of bilinear transformation method for the design of IIR filter?
- f) What is the difference between DSP processor and microprocessor?
- g) Explain symmetric and anti symmetric FIR filters.
- h) What is the time shifting property of DFT?



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- i) What do you understand by DSP Filters?
- j) What is the use of DSP processors?

SECTION-B

2. Consider the sequence shown in the figure below :



The sequence repeats periodically with a period N = 4 for $n \ge 0$ and is zero for n < 0. Find the z-transform of this sequence along with its region of convergence.

3. Consider the filter structure shown in the figure below :



Find the system function and the unit sample response of this system.

- 4. State the properties of DFT. Explain Goertzel algorithm.
- 5. Discuss basic architecture of TMS series of digital signal processors.
- 6. Explain FFT algorithm using decimation in time techniques.



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SECTION-C

- 7. Find the z-transform of each of the following sequences. Also discuss their ROC in details.
 - a) $x(n) = 3\delta(n) + \delta(n-2) + \delta(n+2)$
 - b) x(n) = u(n) u(n 10)
- Consider the following specifications for a bandpass filter : 8.

$$\begin{aligned} \left| H(e^{j\omega}) \right| &\leq 0.01 \qquad 0 \leq \left| \omega \right| \leq 0.2\pi \\ 0.95 \leq \left| H(e^{j\omega}) \right| &\leq 1.05 \qquad 0.3\pi \leq \left| \omega \right| \leq 0.7\pi \\ \left| H(e^{j\omega}) \right| &\leq 0.02 \qquad 0.8\pi \leq \left| \omega \right| \leq \pi \end{aligned}$$

Design a linear phase FIR filter to meet these specifications using a Kaiser window.

- 9. Write a short notes on :
 - a. Linear Convolution
 - www.FirstRanker.com b. Importance of ROC

NOTE : Disclosure of identity by writing mobile number or making passing request on any page of Answer sheet will lead to UMC against the Student.