Roll No

EI/IC-603

B.E. VI Semester

Examination, June 2016

Digital Signal Processing

Time: Three Hours

Maximum Marks: 70

- Note: i) Answer five questions. In each question part A, B, C is compulsory and D part has internal choice.
 - ii) All parts of each question are to be attempted at one place.
 - iii) All questions carry equal marks, out of which part A and B (Max. 50 words) carry 2 marks, part C (Max. 100 words) carry 3 marks, part D (Max. 400 words) carry 7 marks.
 - iv) Except numericals, Derivation, Design and Drawing etc.

Unit - I

- Determine the Fourier transform of u(n).
 - Determine the Fourier transform of $x(n) = \cos \sqrt{2} \pi n$
 - Differentiate between discrete Fourier series and discrete Fourier transform.
 - Explain the linear convolution using DFT.

Discuss about the two dimensional DFT.

Unit - II

- Draw the basic Direct form structure of FIR system.
 - b) Considering a two pole and two zero IIR system draw signal flow graph.
 - Write the names of basic network structures for IIR and FIR systems.
 - Discuss Tellegen's theorem for digital filters and give its applications.

OR

Determine the system function H(z), the impulse response h(n) and the state transition matrix ϕ (n) of the system that generates the Fibonacci sequence. This system is described by the state-space equation.

$$v(n+1) = \begin{bmatrix} 0 & 1 \\ 1 & 1 \end{bmatrix} v(n) + \begin{bmatrix} 0 \\ 1 \end{bmatrix} x(n)$$

$$y(n) = \begin{bmatrix} 1 & 1 \end{bmatrix} v(n) + x(n)$$

Unit - III

[2]

- Name the methods for designing FIR filters.
- Name the techniques for designing IIR filters.
- What is Gibbs phenomenon.
- Discuss how to design a linear phase FIR filters by frequency sampling method.

Discuss bilinear transformation method for designing IIR filter from analog filter.

Unit - IV

- 4. a) Name the efficient computation methods for DFT.
 - b) For an 12 point DFT which efficient computation method should be used and why.
 - c) Write briefly about Goertzel algorithm.
 - Derive the signal flow graph for the N = 8 point, radix 2 decimation in frequency FFT algorithm.

OR

Write short note on chirp-2 transform.

Unit - V

- What are Discrete random signals.
- When does we consider random process ergodic.
- What is the basic principle of spectrum estimation.
- Write short note on the response of linear system to random signals.

OR

Discuss briefly about cross covariance and cross spectrum.