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EC - 603

B.E. VI Semester Examination, June 2015 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 questions 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

- 1. a) Compute N-point DFT of unit sample $\delta(n)$.
 - b) An analog signal is sampled at 15 KHz and 1024 point DFT is computed. Determine the frequency spacing between spectral samples of DFT.
 - c) If x(n) and X(k) be the DFT pair, prove that X(k) obey periodicity property.
 - d) State and prove Symmetry and Periodicity property of Twiddle factor.

OR

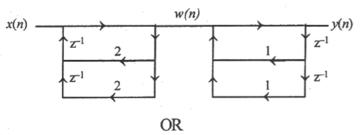
Discuss computational complexity of direct computation of N-point DFT in terms of complex multiplications, addition and real multiplications and additions.

Unit-II

- 2. a) Draw and explain Traversal filter.
 - b) How direct form I structure differs from direct form II structure?
 - c) A difference equation describing a filter is given below, find out filters transfer function also show: - Does this H(z) represent a FIR or IIR filter. Why?

$$y(n) - \frac{3}{4}y(n-1) + \frac{1}{8}y(n-2) = x(n) + \frac{1}{2}x(n-1)$$

 for the flow graph given below write difference equation and system function.



Write Tellegen's theorem for digital filter and its applications.

Unit-III

- . a) How do we map from S plane to Z plane in bilinear transformation?
 - b) How do we obtain causal FIR filters.
 - c) State and prove condition of phase linearity for FIR filter.
 - d) Explain frequency sampling method of FIR filter design.

OR

Make comparison between FIR and IIR filter.

Unit-IV

- a) Explain that Goertzel algorithm is based on linear filtering approach.
 - b) How discrete time and frequency indexes are decompled in chirp Z-transform?
 - c) Prove that in DIT-FFT algorithm input appears in bit reversed order.
 - d) What are the advantages of FFT over DFT's? Also draw butterfly diagram for DIF-FFT algorithm and explain.

OR

Classify FFT algorithm and explain each of them in brief.

Unit-V

- 5. a) What are the limitations of non-parametric method.
 - b) What is energy spectral density?
 - c) Prove that time average autocorrelation function is an estimate of the statistical autocorrelation function.
 - d) Write different properties of autocorrelation function.

OR

Write short note on power spectral estimate.
