

Reg. No.: 920214106015

Question Paper Code: 50444

B.E./B.Tech. DEGREE EXAMINATION, NOVEMBER/DECEMBER 2017 Fifth/Sixth Semester

Electronics and Communication Engineering
EC6502 – PRINCIPLES OF DIGITAL SIGNAL PROCESSING
(Common to: B.E. Biomedical Engineering/Medical Electronics)
(Regulations 2013)

Time: Three Hours

Maximum: 100 Marks

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not extra up of orb beout or PART - A

(10×2=20 Marks)

- 1. What is twiddle factor?
- 2. State and prove periodicity property of DFT.
- 3. List the different types of filters based on frequency response.
- 4. What are the properties of bilinear transformation?
- 5. Write the steps involved in FIR filter design.
- 6. Draw the block diagram representation of FIR system.
- 7. Compare the fixed point and floating point number representations.
- 8. What is meant by finite word length effects in digital system?
- 9. Write the input output relationship for a decimator.
- 10. State the applications of adaptive filtering.

(8-a) x0 0 + (1-a) x0 PART - B

(5×13=65 Marks)

11. a) Find the 8 point DFT of the sequence $x(n) = \{2, 2, 2, 2, 1, 1, 1, 1\}$ using Decimation in Time FFT algorithm.

(OR)

b) Determine the circular convolution of the sequences $x_1(n) = \{1, 2, 3, 1, 1, 2, 3, 1\}$ and $x_2(n) = \{4, 3, 2, 2, 2, 2, 3, 4\}$ using DFT and IDFT.



- 12. a) Enumerate the steps for IIR filter design by impulse invariance with an example. (OR)
 - b) Analyze the design of discrete time IIR filter from analog filter.
- 13. a) Design a FIR filter with the following desired specifications, using Hanning window with N = 5.

$$H_{d}(e^{j\omega}) = \begin{cases} 0, & -\frac{\pi}{4} \le \omega \le \frac{\pi}{4} \\ e^{-j2\omega}, & -\frac{\pi}{4} \le |\omega| \le \pi \end{cases}$$

(OR)

- b) Explain the design procedure of FIR filter using frequency sampling method.
- 14. a) Explain the quantization process and errors introduced due to quantization.

b) i) Explain the characteristics of limit cycle oscillation with respect to the system described by the difference equation:

y(n) = 0.95 y(n-1) + x(n); x(n) = 0 and y(-1) = 13.

ii) Define zero input limit cycle oscillation and explain.

- 15. a) How does the sampling rate increase by an integer factor I? Derive the input-output relationship in both time and frequency domains.

(OR)

b) Discuss in detail about any two applications of adaptive filtering with necessary diagrams.

PART - C

(1×15=15 Marks)

16. a) Obtain the direct form I, direct form II and cascade form realization of the following system function y(n) = 0.1 y(n-1) + 0.2 y(n-2) + 3x(n) + 3.6x(n-1) + 0.6x(n-2).

esx,(n)=(1,2,8,1,1,2,8,1)

b) Convert the given analog transfer function $H(s) = \frac{1}{s+a}$ into digital transfer function by impulse invariant method.