

**APJ ABDUL KALAM TECHNOLOGICAL UNIVERSITY**

**Fifth Semester B.Tech Degree Examination December 2023 (2019 Scheme)**

**Course Code: ECT 303**

**Course Name: DIGITAL SIGNAL PROCESSING**

**Max. Marks: 100 | Duration: 3 Hours (p. 1)**

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**PART A**

***(Answer all questions; each question carries 3 marks) (p. 1)***

- 1. State and prove Parseval's theorem. (p. 1)**
  - 2. Obtain linear convolution of the sequences**  
 **$x(n) = \{1, 2, 3\}$  &  $h(n) = \{-1, -2\}$**  using circular convolution. (p. 1)
  - 3. Find the number of complex multiplications involved in the calculation of a 64-point DFT using (i) direct computation (ii) radix-2 FFT algorithm. (p. 1)**
  - 4. What is twiddle factor? (p. 1)**
  - 5. Derive the mapping between s and z in bilinear transformation. (p. 1)**
  - 6. Given the specification pass band attenuation is 1 dB, stop band attenuation is 30 dB, pass band edge frequency 200 rad/sec and stop band edge frequency 600 rad/sec. Determine the order of the Butterworth Analog filter? (p. 1)**
  - 7. Draw the direct form realization of FIR system. (p. 1)**
  - 8. Why antialiasing filter is used in decimating systems? (p. 1)**
  - 9. What are the different stages in pipelining? (p. 1)**
  - 10. Compare Von Neumann and Harvard architecture. (p. 1)**
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## PART B

(Answer one full question from each module, each question carries 14 marks) (p. 1)

## Module - 1

11. a) Consider the length -12 sequence defined for

$$0 \leq n \leq 11 \text{ is } x(n) = \{3, -1, 2, 4,$$

$$-3, -2, 0, 1, -4, 6, 2, 5\}$$

with a 12-point DFT. Evaluate the following functions of  $X(k)$  without computing DFT: (p. 1)

a.  $X(0)$

b.  $X(6)$

c.  $\sum_{k=0}^{11} X(k)$

d.  $\sum_{k=0}^{11} |X(k)|^2$  (p. 2)

b) State and prove time shifting property of DFT. (6 marks) (p. 2)

OR

12. a) Find the convolution of  $x(n) = \{1, 2, 3, 4, 5, 6, 7, 8, 9\}$  and  $h(n) = \{2, 4, 6\}$  using overlap add method. (7 marks) (p. 2)

b) Find the 4-DFT of the sequence  $\{1, 1, 1, 0\}$  and plot  $|X(K)|$ . (7 marks) (p. 2)

## Module - 2

13. a) Given  $x(n) = (n + 1)$  for  $0 \leq n \leq 7$ . Find  $X(k)$  using DIT-FFT algorithm. (8 marks) (p. 2)

b) Compare DIT and DIF algorithms. (6 marks) (p. 2)

OR

14. a) Explain how a 2N point DFT of a 2N point real-valued sequence can be found by computing a single N point DFT. (6 marks) (p. 2)

b) Find the IDFT of the sequence

$$X(k) = \{4, 1 - j2.414, 0, 1 - j0.414, 0, 1 + j0.414, 0, 1 + j2.414\}$$

using DIF-FFT algorithm. (8 marks) (p. 2)

Module - 3

15. a) Design a Butterworth filter using bilinear transformation. Specifications of desired LPF are:

$$0.9 \leq |H(w)| \leq 1; 0 \leq w \leq \pi/2 \text{ and } |H(w)| \leq 0.2, 3\pi/4 \leq w \leq \pi. T = 1 \text{ sec,}$$

$$A_p = 0.9, A_s = 0.2, w_p = \pi/2, w_s = 3\pi/4. \text{ (8 marks) (p. 2)}$$

b) Derive equations for magnitude and phase responses of FIR filter whose impulse response is symmetric and length N even. (6 marks) (p. 2)

OR

16. a) The desired frequency response of LPF is

$$H_d(w) = e^{-j3w}, |w| \leq 3\pi/4 \text{ and } 0 \text{ otherwise. Determine the frequency response of FIR filter if Hamming window is used. (N=7) (10 marks) (p. 2)}$$

b) Convert the analog filter H(s) given below into a second order Butterworth digital filter using impulse invariance technique.

$$H(s) = \frac{1}{s^2 + \sqrt{2}s + 1} \text{ (4 marks) (p. 2)}$$

**Module - 4**

17. a) When is a cascade form realization preferred in FIR filters? Obtain cascade realization with minimum number of multipliers for the system function

$$H(z) = \left(\frac{1}{2} + z^{-1} + \frac{1}{2} z^{-2}\right)\left(1 + \frac{1}{3} z^{-1} + z^{-2}\right). \quad (7 \text{ marks}) \text{ (p. 2)}$$

(p. 3)

b) How upsampling and downsampling affect the frequency

spectrum of a signal  $x(n)$  with frequency spectrum  $X(e^{j\omega})$ ? Explain the need of low pass filter prior to downsampling. (7 marks) (p. 3)

OR

18. a) What are multirate DSP systems? Give the output of decimation by M system in time domain. (4 marks) (p. 3)

b) Obtain the cascade and parallel form realization for the system

$$y(n) = -0.1y(n-1) + 0.2y(n-2) + 3x(n) + 3.6x(n-1) + 0.6x(n-2). \quad (10 \text{ marks}) \text{ (p. 3)}$$

**Module - 5**

19. a) Explain the effects of coefficient quantization in IIR and FIR filters. (7 marks) (p. 3)

b) What are the main features of a DSP processor? Give the significance of MAC unit in a DSP processor. (7 marks) (p. 3)

OR

20. a) Draw the block diagram of TMS320C67XX and briefly explain the function of each block. (10 marks) (p. 3)

**b) Write a short note on finite word length effects in DSP systems.  
(4 marks) (p. 3)**

