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Name:

APJ ABDUL KALAM TECHNOLOGICAL UNIVERSITY

B.Tech Degree S5 (S, FE) / S5 (PT) (S,FE) Examination June 2024 2015 Scheme

Course Code: EC301

Course Name: DIGITAL SIGNAL PROCESSING

Max. Marks: 100

4

Duration: 3 Hours

PART A

Marks

Answer any two full questions, each carries 15 marks.

1 a) Consider $x(n) = \{1,2,-3,0,1,-1,4,2\}$ with an 8 point DFT X(k). Evaluate the (9) following without finding DFT.

a) X(0) b)
$$\sum_{k=0}^{7} X(k)$$
 c) $\sum_{k=0}^{7} e^{-\frac{j3\pi k}{4}} X(k)$ d) $\sum_{k=0}^{7} |X(k)|^2$ (6)

- b) Using equations, find the 4 point IDFT of $X(k) = \{2, 1+j, 0, 1-j\}$.
- 2 a) Find the output y(n) of a filter whose impulse response is h(n)={1,1,1} and input (10) signal is x(n)={3,-1,0,1,3,2,0,1,2,1} using overlap add method.
 - b) Perform circular convolution of the given sequences using concentric circle (5) method: $x_1(n) = \{1, -1, -2, 3, -1\} \& x_2(n) = \{1, 2, 3\}.$
- 3 a) Find DFT of a sequence $x(n) = \{0, 1, 2, 3, 4, 5, 6, 7\}$ using DIT-FFT algorithm. (11)
 - b) Also compute the number of complex additions & multiplications required for (4) conventional DFT & DIT FFT.

PART B

Answer any two full questions, each carries 15 marks.

a) Determine the low pass filter coefficients h(n) by frequency sampling method for (8) N=7.

 $H_d(e^{j\omega}) = \begin{cases} e^{-j\alpha\omega} & 0 \le |\omega| \le \frac{\pi}{2} \\ 0 & \frac{\pi}{2} \le |\omega| < \pi \end{cases}$

- b) For the analog transfer function, $H(s) = 2/(s^2 + 3s + 2)$, determine H(z) using (7) Impulse Invariance Method. Assume T=1 sec.
- 5 a) Design an ideal LPF with a frequency response given as follows. Find the values (8) of h(n) for N=11.

$$H_d(e^{j\omega}) = 1 \text{ for } -\frac{\pi}{2} \le \omega \le \frac{\pi}{2}$$
$$= 0 \text{ for } \frac{\pi}{2} \le |\omega| \le \pi$$

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- (4)Explain the characteristics of the following window functions: **b**) (ii) Hanning Window (i) Hamming Window If H(z) represents a linear phase FIR filter and has zeros at $z_1 = \frac{1}{\sqrt{2}} + j\frac{1}{\sqrt{2}}$ and $z_2 =$ (3)c) 2. Find the remaining zeros and determine H(z). (10)Butterworth filter. analog given specifications, design an For the a) 6 $|H(j\Omega)| \le 0.2$, for 0.4 $\pi \le \Omega \le \pi$ $0.9 \le |H(j\Omega)| \le 1$, for $0 \le \Omega \le 0.2\pi$ &
 - b) Explain pre-warping in bilinear transformation.

PART C

(5)

(10)

(6)

Answer any two full questions, each carries 20 marks.

7 a) Explain transposed form of a filter structure. Obtain the transposed direct form II (10) for the given IIR filter.

$$H(z) = \frac{1 + 2z^{-1} + z^{-2}}{1 - \frac{3}{4}z^{-1} + \frac{1}{8}z^{-2}}$$

b) Obtain the following realizations for the given FIR filter.

(i) Direct form realization with minimum number of multipliers

(ii) Cascade realization with minimum number of multipliers

$$H(z) = \left(\frac{1}{2} + z^{-1} + \frac{1}{2}z^{-2}\right)\left(1 + \frac{1}{3}z^{-1} + z^{-2}\right)$$

8 a)

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

Obtain the cascade realization for the following system:

- b) Realize the system with difference equation y(n) = -0.1 y(n (14))1) + 0.72 y(n - 2) + 0.7x(n) - 0.252 x(n - 2) using direct form I, direct form II and parallel form.
- 9 a) With an example, illustrate the error introduced by truncation and rounding in (6) floating point representation of numbers.
 - b) How is a floating-point number represented in a processor? Explain the operations (7) of addition and multiplication of two floating point numbers with examples.
 - c) Explain ADC quantization noise. Derive the variance of quantization noise in ADC (7) with step size Δ .
