Paper / Subject Code: 32222 / Discrete Time Signal Processing

June 5, 2024 02:30 pm - 05:30 pm 1T01035 - T.E. (Electronics and Telecommunication)(SEM-V)(Choice Base Credit Grading System) (R- 19) (C Scheme) / 32222 - Discrete Time Signal Processing QP CODE: 10055692

Duration: 3hrs [Max Marks:80]

- N.B.: (1) Question No 1 is Compulsory.
 - (2) Attempt any three questions out of the remaining five.
 - (3) All questions carry equal marks.
 - (4) Assume suitable data, if required and state it clearly.
- 1 Attempt any **FOUR**

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a Obtain the linear phase realization of the system function

$$H(z) = \frac{1}{2} + \frac{1}{3}z^{-1} + z^{-2} + \frac{1}{4}z^{-3} + z^{-4} + \frac{1}{3}z^{-5} + \frac{1}{2}z^{-6}$$

- b Discuss the properties of Butterworth and type-I Chebyshev filter
- c Given H(Z), compute the truncated H'(z) with coefficients represented by 4-bit word length. $H(z) = \frac{1}{z 0.752352}$
- d Find the IDIF-FFT for a given sequence $X(k) = \{26, -2+2j, -2, -2-2j\}$.
- e State the condition for stability and causality of analog and digital filter
- 2 a Design a lowpass filter for the following specification [10]

 Cutoff frequency=500 Hz; sampling frequency = 2000 Hz; order of the filter

 N=10.Use Hamming window to get the impulse response.
 - b Determine the output response y(n) if h(n)= (1,1,1), x(n)= (1,2,3,1) by using linear convolution, circular convolution & circular convolution with zero padding.
- 3 a Find the DFT of the sequence $x[n] = \{1, 2, 3, 4, 4, 3, 2, 1\}$ using radix-2 [10] decimation in time FFT algorithm.
 - b For the analog transfer function $(s) = \frac{2}{s^2 + 3s + 2}$. Determine H(z) using impulse invariant transformation if (a) T=1 second and (b) T=0.1 second.

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4 a Determine the impulse response h(n) of a filter having desired response=7, use [10] frequency sampling approach

$$Hd(w) = egin{cases} e^{-j3\omega} & 0 \leq |\omega| \leq rac{\pi}{2} \\ 0 & rac{\pi}{2} \leq |\omega| \leq \pi \end{cases}$$

- b Derive the equation for steady state noise power due to input quantization. [10]
- 5 a Design a digital Butterworth low pass filter that satisfies the following constraint using BLT. Assume T = 1 sec

$$0.707 \le |H(w)| \le 1$$
; for $0 \le \omega \le 0.5\pi$
 $|H(w)| \le 0.2$; for $0.75\pi \le \omega \le \pi$

b Determine the zeros of the following FIR systems and identify whether the following system is minimum phase, maximum phase, mixed phase. Also comment on stability.

(i)
$$H_1(z) = 6 + z^{-1} + 6z^{-2}$$

(ii)
$$H_2(z) = 1 - z^{-1} - 6z^{-2}$$

(iii)
$$H_3(z) = 1 - \frac{5}{2}z^{-1} - \frac{3}{2}z^{-2}$$

(iv)
$$H_4(z) = 1 - \frac{5}{2}z^{-1} - \frac{2}{3}z^{-2}$$

6 a Describe echo cancellation and sub-band coding of speech signal. [10]

[10]

b Obtain the cascade and parallel form realization of the system

$$Y(n) = -0.1y(n-1) + 0.2y(n-2) + 3x(n) + 3.6x(n-1) + 0.6x(n-2)$$