(07 Marks)



Fifth Semester B.E. Degree Examination, Jan./Feb. 2021 Digital Signal Processing

Time: 3 hrs. Max. Marks: 100

Note: Answer any FIVE full questions, choosing ONE full question from each module.

Module-1

- 1 a. Show that the multiplication of two DFT's leads to circular convolution of the corresponding time sequences. (08 Marks)
 - b. Compute the N point OFT's of the signals:

i)
$$x(n) = \begin{cases} 1, & 0 \le n \le \frac{N}{2} - 1 \\ 0, & \frac{N}{2} \le n \le N - 1 \end{cases}$$

- ii) $x(n) = \cos \frac{2\pi}{N} k_0 n$, $0 \le n \le N-1$.
- c. Given $x(n) = \{1, 2, 3, 4\}$, find y(n), if $y(k) = x((k-2))_4$. (05 Marks)

OR

- 2 a. State and prove the Circular time shift property of DFT. (06 Marks)
 - b. Determine the circular convolution of $x_1(n) = \{1, 2, 3, -1\}$ and $x_2(n) = \{4, 3, 2, -2\}$, using

Time domain formula. Verify the result using Frequency domain approach. (09 Marks)

- c. For the sequence $x(n) = \{-1, 2, 3, 0, -4, 1, 2, -3\}$, Calculate
 - i) $\sum_{k=0}^{7} x(k)$ and ii) $\sum_{k=0}^{7} |x(k)|^2$, without computing the DFT. (05 Marks)

Module-2

3 a. Write the computational procedure to find the filtered output using Overlap Add method.

Find the 8 – point DFT of the sequence $x(n) = \{-1, 0, 2, 3, -4, -2, 0, 5\}$, using radix – 2

- b. Find the 8 point DF1 of the sequence $x(n) = \{-1, 0, 2, 3, -4, -2, 0, 5\}$, using radix 2 DIT FFT algorithm. (09 Marks)
- c. Compare the complex additions and complex multiplications for the direct computation of DFT versus the FFT algorithm for N = 128. (04 Marks)

OR

- a. Derive the radix 2 DIF FFT algorithm and draw the signal flow graph for N = 8.
 Comment on the number of computations required to find N point DFT. (07 Marks)
 - b. Using Overlap save method, find the output of a filter whose impulse response $h(n) = \{1, -2, 3\}$ and input $x(n) = \{2, 3, -1, 0, 5, 2, -3, 1\}$.

Use 6 – point circular convolution. (09 Marks)

c. Given $X(k) = \{1, j4, 1, -j4\}$, find x(n) using radix -2 DIT -FFT algorithm. (04 Marks)

Any revealing of identification, appeal to evaluator and /or equations written eg, 42+8 = 50, will be treated as malpractice Note: 1. On completing your answers, compulsorily draw diagonal cross lines on the remaining blank pages

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Module-3

Design an FIR filter for the following desired frequency response

$$H_{d}(w) = \begin{cases} e^{-j3w}, & \text{if } |w| \leq \frac{\pi}{4} \\ 0, & \text{if } |w| > \frac{\pi}{4} \end{cases}.$$

Use the Hamming window function, obtain the frequency response of the designed FIR (10 Marks)

b. For the System function $H(z) = 1 + 2.8z^{-1} + 3.4z^{-2} + 1.7z^{-3} + 0.4z^{-4}$. Obtain the Lattice coefficients and sketch the Lattice structure. (10 Marks)

Find the Impulse response of an FIR filter with the following desired frequency response,

$$H_{d}(w) = \begin{cases} 0 & ; & \text{if } |w| \leq \frac{\pi}{6} \\ e^{-j4w} & ; & \text{if } |w| > \frac{\pi}{6} \end{cases}.$$

Use Rectangular window function. Draw the direct form structure for the designed filter.

b. Consider an FIR Lattice filter coefficients $K_1 = 0.65$, $K_2 = 0.5$, $K_3 = 0.9$. Find its impulse response and draw the direct form structure. (10 Marks)

Module-4

- a. Define the First order analog low pass filter prototype. How this prototype is transformed into a different filter types. (05 Marks)
 - b. Design a Second order digital low pass Butterworth filter with a cutoff frequency of 3.4 kHz at a sampling frequency of 8000Hz. Draw the direct Form – II structure of this filter. Use Bilinear transformation. (10 Marks)
 - c. Discuss the general mapping properties of bilinear transformation and show the mapping between the S – plane and the the Z – plane. (05 Marks)

- a. Define the Normalized low pass prototype function of Butterworth filter and derive the expression for the filter order.
 - b. Using Bilinear transformation, design a digital low pass Butterworth fitler with the following specifications: Sampling frequency: 8000Hz, 3 dB attenuation at 1.5 kHz. 10 dB stop band attenuation at 3kHz. (10 Marks)

c. Realize the following digital filter using direct Form – II
$$H(z) = \frac{0.7 + 1.4z^{-1} + 0.7z^{-2} + 0.5z^{-3}}{1 + 1.3z^{-1} + 0.5z^{-2} + 0.7z^{-3} + 0.3z^{-4}}.$$
 (05 Marks)

- a. With a neat diagram, explain the Harvard architecture used in DS processors. (06 Marks)
 - Illustrate the operation of circular buffers used for address generation in DS processors. (07 Marks)
 - Convert the following decimal numbers into the floating point representation 0.640492×2^{-2} ii) -0.638454×2^{5} .

(07 Marks)

OR

- With a neat diagram, explain the basic architecture of TMS320C54X family DS processors. (10 Marks)
 - Describe the IEEE single precision floating point format used in DS processors. (05 Marks)
 - Find the signed Q 15 representation for the decimal number 0.560123. (05 Marks)

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