CRASH COURSE

Fifth Semester B.E. Degree Examination, May 2017 Digital Signal Processing

Time: 3 hrs.

Max. Marks: 100

Note: Answer FIVE full questions, selecting at least TWO questions from each part.

PART - A

- 1 a. Consider a finite length sequence $x(n) = \delta(n) + 2\delta(n-5)$. Find
 - i) The 10 point DFT of x(n)

 $j2\pi 2k \\$

- ii) Find the sequence that has a DFT $y(k) = e^{-10} x(k)$
- iii) Find the 10 point sequence y(n) that has DFT y(k) = x(k) w(k), where x(k) is the 10 point DFT of x(n) and y(k) is the 10 point DFT of y(n) given by y(n)

 $\begin{cases} 1 & 0 \le n \le 6 \\ 0 & \text{otherwise} \end{cases}$

(12 Marks)

- b. Find the DFT of the sequence $x(n) = 0.5^n u(n)$ for $0 \le n \le 3$ by evaluating $x(n) = a^n$ for $0 \le n \le N-1$.

3, 2, 1, -1, -2, -3, 5, 6, -1, 2, 0, 2, 1} find the output using overlap add method, assuming the length of block as 7. (12 Marks)

- b. State and prove the following properties i) Linearity ii) Circular time shift. (08 Marks)
- 3 a. What are FFT algorithms? Prove the i) symmetry and ii) periodicity property of the twiddle factor W_{N_c} (07 Marks)
 - b. What are the advantages of FFT algorithm?

(04 Marks)

- c. Use the 8 point radix- 2 DIT FFT algorithm to find the DFT of the sequence $X(n) = \{0.707, 1, 0.707, 0, -0.707, -1, -0.707, 0\}.$ (09 Marks)
- 4 a. Develop decimation in frequency (DIF)-FFT algorithm with all necessary steps and neat signal flow diagram used in computing N-point DFT, x(k) of a N point sequence (x(n).

 (12 Marks)
 - b. What is Goertzel algorithm? For sequence x(n) = (2, 0, 2, 0) determine x(2) using Goertzel algorithm. Assume initial conditions are zero. (08 Marks)

PART - B

5 a. Design a Chebyshev I filter to meet the following specifications

(12 Marks)

i) Passband ripple :ii) Passband edge :

 $\leq 2dB$

- ii) Passband edge : 1 rad/seciii) Stopband attenuation : ≥ 20dB
- iv) Stopband edge : 1.3 rad/sec
- b. Derive the expression for order and cutoff frequency of a Butterworth low pass filter.

(08 Marks)

6 a. A liner time invariant digital IIR filter is specified by the following transfer function

$$H(z) = \frac{(z-1)(z-2)(z+1)z}{\left[z - \left(\frac{1}{2} + j\frac{1}{2}\right)\right]\left[z - \left(\frac{1}{2} - j\frac{1}{2}\right)\right]\left[z - j\frac{1}{4}\right]\left[z + j\frac{1}{4}\right]}$$

Realize the system in the following forms i) direct form I ii) direct form II (12 Marks)

- b. Realize an FIR filter with impulse response h(n) given by h(n) = $\left(\frac{1}{2}\right)^n \left[u(n) u(n-u)\right]$ using direct form I. (08 Marks)
- 7 a. Obtain the 10 coefficients of an FIR filter to meet the specifications given below using Hamming window method (12 Marks)

Passband edge frequency : 1.5 KHz
Stopband edge frequency : 2 KHz
Minimum stopband attenuation : 50 dB
Sampling frequency : 8 KHz

- b. Derive the frequency response of a symmetric FIR low pass filter for both N even and N odd. (08 Marks)
- 8 a. Show that the bilinear transformation maps
 - i) The $i\Omega$ axis in s-plane onto the unit circle, |z| = 1

ii) The left half S plane, Re(S) < 0 inside the unit circle, |z| < 1. (10 Marks)

b. A digital lowpass filter is required to meet the following specifications

 $20 \log |H(w)|w = 0.2\pi \ge -1.9328 \text{ db}$

 $20 \log |H(w)|w = 0.6\pi \le -13.9794 \text{ dB}$

The filter must have a maximally flat frequency response. Find H(z) to meet the above specifications using impulse invariant transformation. (10 Marks)

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