T.E.(ELECTRONICS & TELE-COMMN.)(Sem VI) (CBSGS) / 37002 - DISCRETE TIME SIGNAL PROCESSING

Q.P. CODE: 37019

Time: 3 Hours   **Marks: 80**

N.B: (1) Questions No.1 is compulsory.
      (2) Attempt any three questions out of remaining five questions.
      (3) Assume suitable data if required.
      (4) Figures to the right indicate full marks.

Q 1.Solve any four

    a) Determine the zeros of the following systems and indicate whether the system is minimum, maximum or mixed phase.

        1) $H_1(z) = 6+z^{-1}+6z^{-2}$
        2) $H_2(z)= 1-z^{-1}-6z^{-2}$ 5

   b) Define group delay and phase delay.   5

   c) Compare FIR and IIR filters   5

   d) What is frequency warping in bilinear transformation.   5

Q2 a) Compute DFT of sequence x(n)={ 2,1,2,1,1,2,1,2 } using DIT-FFT algorithm.   10

   b) A low pass filter is to be designed with following desired frequency response.

       $Hd(e^{jw})=e^{-j2w}$ $-\frac{\pi}{4}\le w \le \frac{\pi}{4}$

           $=0$ $\frac{\pi}{4}< w \le \pi$

   Determine the filter coefficients $h_d(n)$ if the window function is defined as

   w(n)=1   0≤n≤4

    =0   otherwise

Also determine the frequency response $H(e^{jw})$ of the designed filter.   10

Q 3 a)The transfer function for discrete time system is given as

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

i) Draw Direct Form I and Form II realization

ii)Draw cascaded and parallel form realization   10

b) Explain subband coding of speech signal as a application of multirate signal processing. 10

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Q4 a) Develop composite radix DITFFT flow graph for N=6=2×3.   10

 b) Design a digital Butterworth filter that satisfies following constraints using bilinear transformation method. Assume Ts=1s.

$$0.9 \leq \left| H\left(e^{jw}\right) \right| \leq 1 \qquad 0 \leq w \leq \frac{\pi}{2}$$

$$\left| H\left(e^{jw}\right) \right| \leq 0.2 \qquad \frac{3\pi}{4} \leq w \leq \pi \qquad\qquad 10$$

Q 5 a) Show the mapping from S plane to Z plane using impulse invariant method. Explain its limitations. Using this method determine H(z) if

$H(s)=\dfrac{10}{(s+5)(s+2)}$ if Ts=0.2s.   10

b) If x(n)={ 1,2,3,} and h(n)= {1,0}

        1) Find linear convolution using circular convolution
        2) Find circular convolution using DFT-IDFT.   10

Q6   Write short notes on following,

        a. Musical Sound Processing.   07

        b. Dual tone multi frequency signal detection.   06
        c. Subband Coding of Speech signals.   07

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