#### ADAMAS UNIVERSITY **END-SEMESTER EXAMINATION: MAY 2021** (Academic Session: 2020 – 21) Name of the Program: B.Tech. VIII Semester: Advanced Digital Signal Processing Paper Title: **Paper Code:** EEC61102 40 **Maximum Marks:** Time duration: 3 Hours **Total No of questions:** 08 **Total No of** 02 Pages: Note: 1. Follow all the Instructions given on the cover page of the Answer Booklet Strictly. 2. All parts of a Question should be answered consecutively. 3. Each answer should start from a fresh page.

# Answer all the Groups Group A

4. Assumptions made if any, should be stated clearly at the beginning of

Answer all the questions of the following

 $5 \times 1 = 5$ 

- 1. a) What do you mean by an "Adaptive Filter"?
  - b) Show that,  $W_N^{2kn} = W_{N/2}^{kn}$
  - c) How many ways you can digitize an analog filter?

your answer.

- d) Write down the different application of advanced digital signal processing.
- e) Given the specification

passband attenuation,  $\alpha_p$ =1 dB, stopband attenuation,  $\alpha_s$ = 30 dB, passband frequency,  $\Omega_p$ = 200 rad/s and stopband frequency,  $\Omega_s$ = 600 rad/s. Determine the order of the filter.

#### **GROUP-B**

Answer any three of the following

 $3 \times 5 = 15$ 

- **2.** Draw and explain the block diagram of a digital communication system used for echo cancellation in a modem.
- **3.** Explain the process of digital watermarking using block diagram.
- **4.** Write down short notes on Adaptive Direct-Form filter using LMS (Least Mean Squared) algorithm.
- 5. Find the linear convolution using circular convolution for two sequences  $x(n)=\{1,2,-1,2,3,-2,-3,-1,1,1,2,-1\}$  and  $h(n)=\{1,2\}$  using Overlap add method.

Or

Determine the frequency response of FIR filter defined by  $y(n)=0.25 \ x(n)+x(n-1)+0.25 \ x(n-2)$ . Calculate the phase and group delay.

### **GROUP-C**

## Answer any two of the following

 $2 \times 10 = 20$ 

- **6.** a) Draw and explain the block diagram model for the generation of a speech signal.
  - b) How source encoding is possible for a speech signal? Explain with proper block diagram.

[5+5]

- 7. Compute 8-point DFT of the following sequence  $x(n) = \{0.5, 0.5, 0.5, 0.5, 0, 0, 0, 0, 0\}$  using DIT-FFT algorithm.
- **8.** a) Draw and explain state space Kilman filter.
  - b) Write the necessary equations for state process model and observation model.

[7+3]