

(A Constituent College of Somaiya Vidyavihar University) **Department of Electronics Engineering** 



| Course Name:                | Digital Signal & Image Processing<br>Laboratory | Semester:    | VI          |
|-----------------------------|---|--------------|-------------|
| <b>Date of Performance:</b> | 20 / 03 / 2025                                  | Batch No.:   | B - 2       |
| Faculty Name:               | Dr. Om Goswami                                  | Roll No.:    | 16014022050 |
| Faculty Sign & Date:        |   | Grade/Marks: | / 20        |

# **Experiment No: 5**

**Title:** Implementation of FIR/IIR filters

## **Objective:**

To design and implement a Finite Impulse Response (FIR) and Infinite Impulse Response filters

#### COs to be achieved:

CO2: Design different filters in digital domain.

Materials Required: Python/MATLAB software

**Books/ Journals/ Websites referred:** 

- 1. A. Nagoor Kani "Digital Signal Processing", 2<sup>nd</sup> Edition, TMH Education.
- 2. Alan V. Oppenheim and Ronald W. Schafer, "Discrete-Time Signal Processing."
- 3. MATLAB Documentation: https://www.mathworks.com/help/matlab/

## Theory:

A Finite Impulse Response (FIR) filter is a type of digital filter where the impulse response is of finite duration because it settles to zero in a finite number of sample intervals. The output of an FIR filter is given by the convolution of the input signal with the filter coefficients (impulse response).

#### **Mathematical Representation:**

FIR Filters: For an input sequence x[n], the output sequence y[n] of an FIR filter with coefficients h[k]is:

$$y(n) = \sum_{k=0}^N b_k x(n-k)$$

Where:

y(n) is the output signal,

x(n) is the input signal,

b<sub>k</sub> are the filter coefficients (impulse response),

N is the filter order (determines how many past inputs are used).



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#### Properties of FIR Filters

- Linear phase response (if coefficients are symmetric or anti-symmetric).
- Always stable (since there are no feedback elements).
- Can be designed to have an exact linear-phase response.

Infinite Impulse Response (IIR) filters are a class of digital filters used in signal processing. Unlike Finite Impulse Response (FIR) filters, IIR filters have feedback, which allows them to have an infinite duration response to an impulse input.

#### **Characteristics of IIR Filters:**

- 1. Recursive Structure: IIR filters use both past and present input values as well as past output values to compute the current output.
- **2. Infinite Impulse Response:** Since the filter has feedback, the impulse response theoretically extends indefinitely.
- **3.** Efficient Implementation: Compared to FIR filters, IIR filters typically require fewer coefficients to achieve a given frequency response, making them computationally efficient.
- **4. Potential for Stability Issues:** Due to feedback, an improperly designed IIR filter can become unstable if the poles of its transfer function lie outside the unit circle in the Z-domain.

## **Mathematical Representation:**

An IIR filter is described by the difference equation:

$$H(z) = rac{\sum_{k=0}^{N} b_k z^{-k}}{1 + \sum_{k=1}^{M} a_k z^{-k}}$$

where:

y(n) is the output signal,

x(n) is the input signal,

ak and bk are filter coefficients,

M and N are the number of feedback and feedforward coefficients, respectively.



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| Stepwise-Procedure:  FIR Filters: <a href="https://dsp-iitkgp.vlabs.ac.in/exp/fir-lowpass-highpass/">https://dsp-iitkgp.vlabs.ac.in/exp/fir-lowpass-highpass/</a> Screenshots:  |
|---|
| ≡   |
| 1. Select Firk from Type 2. Select the sampling frequency (F <sub>a</sub> ) and centre frequency (F <sub>c</sub> ) of the input signal. 3. Put the value of Firk filter order from FIR croter tab select filter type from Calculation sinc lowpass or sinc highpass and attenuation(dB) 4. Click on the "Calculate" button to observe the Magnitude, Group delay, Impulse Response and simulation on plot Note:  o Make sure FIR order in range of 1 to 100   |
| Type: FIR v Fs. 1000 Fc 100 Fc2 200   |
| Biquad cascades: 3  Transform:   bilinear   |
| FIR order: 100 Calculation Sinc Lowpass V Attenuation (dB): 100   |
| calculate Magnitude Percent   |
| 39.71 Hz 99.99 %  |
| 90.00 % -<br>80.00 % -<br>\$\overline{\pi}\$ 70.00 % -<br>\$\overline{\pi}\$ 60.00 % -  |
|   |
| FIR order: 100 Calculation Sinc Lowpass ✓ Attenuation (dB): 100   |
| calculate  Magnitude Percent  |
| 464.95 Hz 0.00 %  |
| 100.00 %<br>90.00 %   |
| 80.00%  |
| With the state of |
| 40.00 % · 40.00 % ·   |
| 30.00 %   |
| 20.00 % ·<br>10.00 % ·  |
| 0 50 100 150 200 250 300 350 400 450  |
| Frequency [Hz]  |
| Magnitude dB  |
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Academic Year: 2024-25 Roll No.: 16014022050 DS&IP Semester: VI



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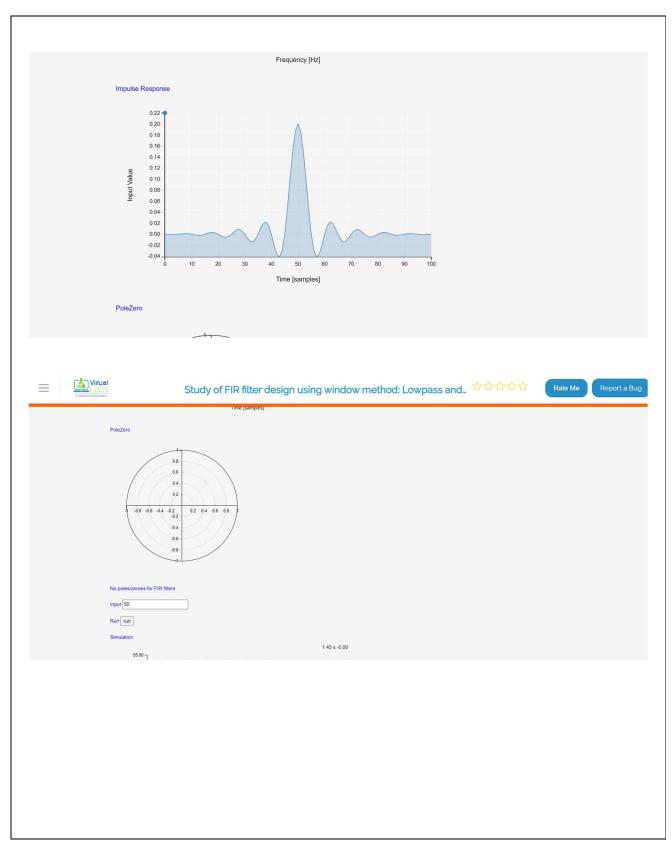
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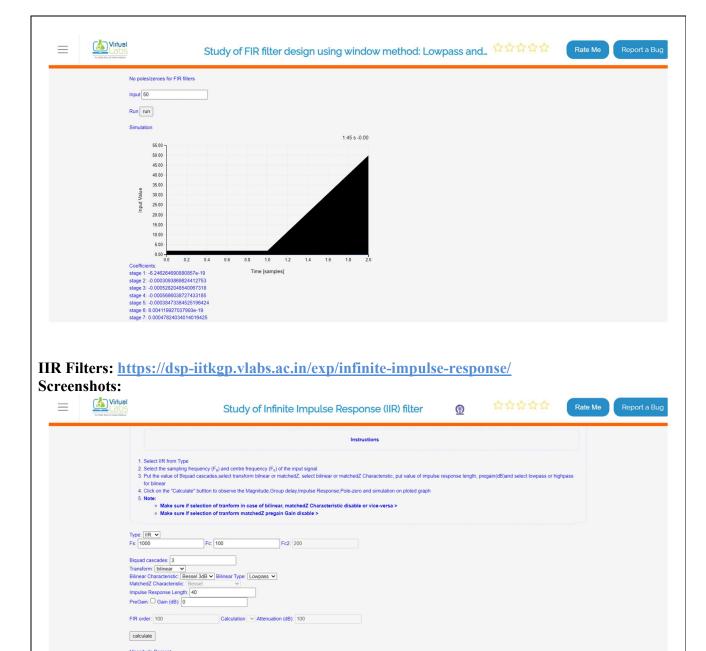






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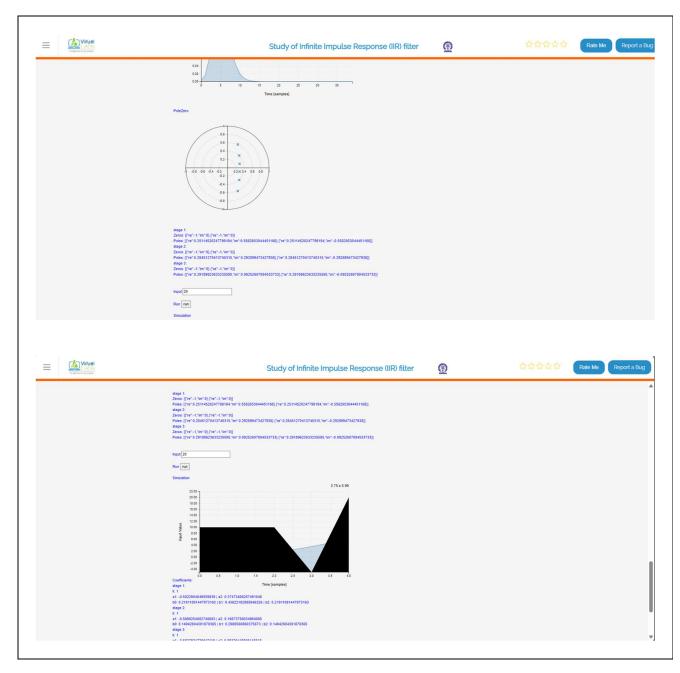






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#### **Conclusion:**

In this experiment, we designed and implemented FIR and IIR filters using the virtual lab. It helped us understand the filtering process and the difference between FIR and IIR filters in signal processing.

#### **Post Lab Question:**

## 1. What is the role of the window function in FIR filter design?

The window function is used in FIR filter design to control the frequency response of the filter. When designing an FIR filter using the Fourier series method, truncation of the impulse response can introduce unwanted ripples in both the passband and stopband. The window function smooths the transition and reduces these ripples by tapering the impulse response at the edges, thereby improving the filter's performance. Different window functions (e.g., Hamming, Hanning, Blackman) offer trade-offs between main lobe width and side lobe attenuation.

## 2. How does the filter order affect the filter's performance?

The filter order determines the number of coefficients in the FIR filter and directly affects its performance:

- Higher Order: Improves frequency selectivity, reduces transition bandwidth, and achieves better attenuation in the stopband. However, it increases computational complexity.
- Lower Order: Reduces computational cost but results in a wider transition band and poorer stopband attenuation.

In general, a higher-order filter provides better performance but at the expense of increased processing time and hardware complexity.

## 3. Why are FIR filters always stable?

FIR filters are always stable because they do not use feedback in their implementation. Their impulse response is finite, meaning they only rely on present and past input values without recursive elements. As a result, there is no possibility of an unstable oscillatory response, making FIR filters inherently stable under all conditions.

## 4. What is the difference between FIR and IIR filters?

The main difference between FIR (Finite Impulse Response) and IIR (Infinite Impulse Response) filters lies in their impulse response and structure.

- FIR filters have a finite duration impulse response because they depend only on the current and past input values. They are always stable and have linear phase characteristics, making them ideal for applications where phase accuracy is important.
- IIR filters have an impulse response that lasts indefinitely because they use both past input and past output values (feedback). They can achieve a desired filter response with a lower order than FIR filters but may become unstable if not designed



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carefully. IIR filters typically have a non-linear phase response.

**Signature of faculty in-charge with Date:**