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# Assignment-1

## **Analog Electronics**

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- 1) a) Generate a square wavefrom (approximate) using sinusoildal waveforms with different frequencies. (start from 5 kHz)
  - b) Pass this square wavefrom with an amplifier with a gain A(w) and the following frequency response
    - i) (upto 10 kHz A(w) = 1)
    - ii) (From 10 kHz to 50 kHz A(w) = 5)
    - iii) (From 50 kHz to 200 kHz A(w) = 2)
    - iv) (From 200 kHz to 1 MHz A(w) = 1)
    - v) (From 1 MHz to 2 MHz A(w) = 0.3)
    - vi) (From 2 MHz and above A(w) = 0)
  - c) Pass the square waveform generated in (a) with an ideal low-pass filter with cut-off frequency = 8 kHz. Get the output waveform.

Solution: Here the MATLAB to generate the above waveforms

```
% Define time steps and initialize signal arrays
t = 0:0.001:1; % Create a time vector with 0.001s steps from 0 to 1 second
y = zeros(size(t)); % Initialize two empty arrays for storing signals
y1 = zeros(size(t));
% Define constants
k = 2 * pi; % Constant used for calculating sine wave frequencies
% Loop through harmonics (odd multiples of 5)
for i = 1:2:100
    % Define amplitude based on harmonic frequency
    if 5 * i >= 5 \&\& 5 * i < 10
        A = 1; % Amplitude for harmonics between 5 and 10 Hz
    elseif 5 * i >= 10 \&\& 5 * i < 50
        A = 5; % Amplitude for harmonics between 10 and 50 Hz
    elseif 5 * i >= 50 \&\& 5 * i < 200
        A = 2; % Amplitude for harmonics between 50 and 200 Hz
    elseif 5 * i >= 200 \&\& 5 * i < 1000
        A = 1; % Amplitude for harmonics between 200 and 1000 Hz
    elseif 5 * i >= 1000 & 5 * i < 2000
        A = 0.3; % Amplitude for harmonics between 1000 and 2000 Hz
    elseif 5 * i >= 2000
        A = 0; % No contribution for harmonics at or above 2000 Hz
    end
    % Add sine wave component for harmonics within 5 – 100 Hz range
    if 5 * i >= 5 && 5 * i <= 100
```

```
y = y + (4/(pi * i)) * sin(k * 5 * i * t); % Add weighted sine wave to original signal
    end
    % Add scaled sine wave component for all harmonics to amplified signal
    y1 = y1 + A * (4 / (\mathbf{pi} * i)) * \mathbf{sin}(k * 5 * i * t); \% Add weighted and scaled sine wave to
        amplified signal
end
% --- Implementing Low-Pass Filter ---
% Get the Fast Fourier Transform of the amplified signal
Y1 = \mathbf{fft}(y1);
% Define frequency domain and create filter mask
f = k * (1:length(Y1)) / (2 * pi); % Calculate frequencies from the FFT data
filter mask = ones(size(f)); % Create a mask with all frequencies initially passed
% Set stopband mask to attenuate frequencies above 8 Hz
filter mask(f > 8) = 0; % Set all frequencies above 8 Hz to zero in the mask
% Apply filter and inverse transform to get filtered signal
Y2 = filter mask .* Y1; % Apply the filter mask to the frequency spectrum
y2 = ifft(Y2); % Convert the filtered frequency spectrum back to time domain
% --- Plotting the Signals ---
% Figure 1: Original Signal
figure;
plot(t, y, 'b');
xlabel('Time_(s)');
ylabel('Amplitude');
title('Original_Signal');
grid on;
% Figure 2: Amplified Signal
figure;
plot(t, y1, 'g');
xlabel('Time_(s)');
ylabel('Amplitude');
title('Amplified_Signal');
grid on:
% Figure 3: Filtered Signal
figure;
plot(t, y2, 'r');
xlabel('Time_(s)');
ylabel('Amplitude');
title('Filtered_Signal_(Figure_3)');
grid on;
```

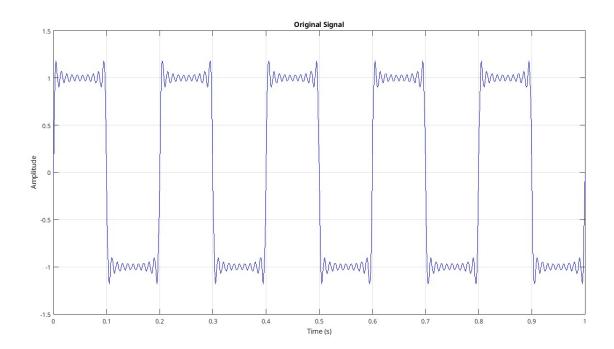


Fig. 1. The i/p signal generated for 100 harmonics,

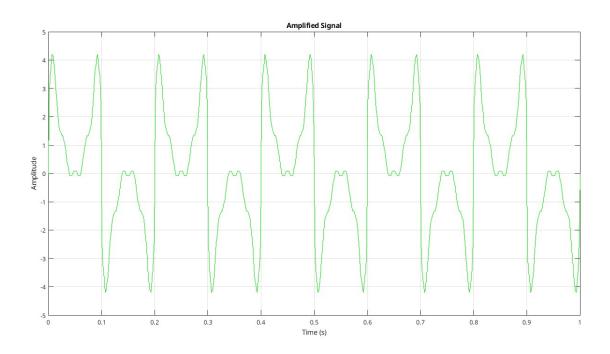


Fig. 1. The amplified i/p signal with variable amplification at different frequencies,

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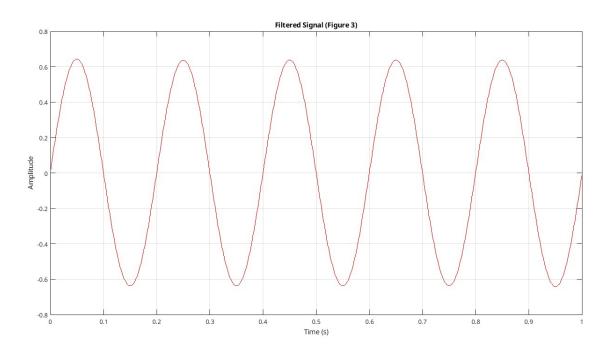


Fig. 1. The filtered ampliplified i/p signal with cutoff at 8khz

### Conclusion:

We created a square wave by putting together different sinusoidal waves, starting at 5 kHz. We then ran this square wave through an amplifier that changed how much it boosted or reduced different frequency ranges. After that, we used a filter to keep only the parts of the wave below 8 kHz and cut out the rest. The end result was a modified waveform, showing how amplification and filtering can alter the original square wave.

2) A nonlinear device is with following input/output characteristics:

$$I_{out} = \beta V_{in}^2 \tag{1}$$

Consider  $\beta = 10 \text{mA/V2}$ , and  $V_{in} = 1 \text{V}$ . Find out DC operating point corrospondingly. Find out the range of incremental change in Vin for that the device still considered as a linear one with less than (i) 1%, (ii) 5% non-linearilty.

#### **Solution:**

$$= \frac{0.02}{0.99} = +0.0202 \text{ Volts}$$

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$$\Rightarrow -0.02 \times 1 = 1.01 \text{ DVin}$$

DVin,1% € { - 0.0198, +0.02027 V

$$\pm 0.05 = \frac{\Delta V_{in}}{2V + \Delta V_{in}} = \pm 0.05 = \frac{\Delta V_{in}}{2 + \Delta V_{in}}$$