EE 511 - Homework 02

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Assuming the following waveform

```
\left\{egin{array}{l} 5sin(\omega t+\pi/6) \;\; t<0.1 \ 10sin(\omega t+\pi/6) \;\; t\geq0.1 \end{array}
ight.
```

Given that, f = 60 Hz, and sampling frequency is 12 samples per cycle.

```
In [1]: ## IMPORT PACKAGES
import matplotlib.pyplot as plt
import matplotlib.colors as mcolors
import numpy as np
```

Plotting the input signal (no harmonics):

```
In [3]: ## PLOTTING INPUT SIGNAL
    time = np.arange(start_time,end_time,dT)
    voltage1 = []

for t in range(0,len(time)):

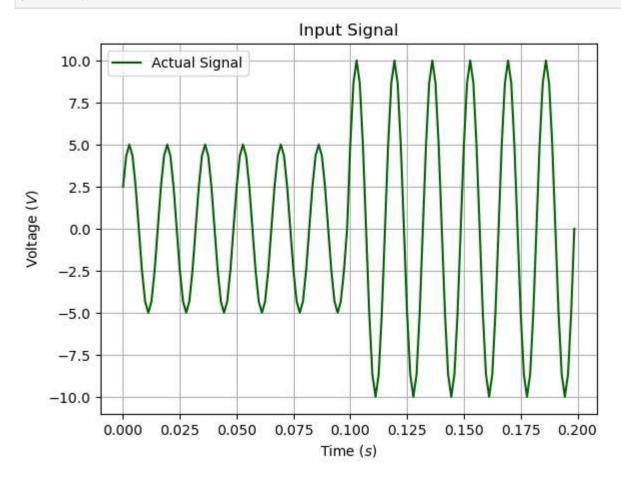
    if time[t] < 0.1:
        v = A1 * np.sin(w * time[t] + np.pi/6) # first signal
    else:
        v = A2 * np.sin(w * time[t] + np.pi/6) # second signal

    voltage1.append(v)

voltage1 = np.array(voltage1)

plt.plot(time,voltage1,mcolors.CSS4_COLORS['darkgreen'])
plt.title("Input Signal")
plt.xlabel("Time $(s)$")
plt.ylabel("Voltage $(V)$")
plt.ylabel("Noltage $(V)$")
plt.legend(["Actual Signal"])</pre>
```

plt.grid()
plt.show()



a) Estimate the amplitude of the signal using full-cycle Fourier algorithm

Fourier algorithm function:

```
In [4]: # FOURIER ALGORITHM
        def FourierAlgorithm(signal, frequency, sample):
            y = signal
            n = y.size
            amplitude = []
            angle = []
            f = frequency
            N = sample
            dT = 1/(f*N)
            t = np.arange(start_time,end_time,dT)
            time = t[N - 1:n]
            # WEIGHT VECTOR
            w1 = np.cos(np.arange(0, N) * 2 * np.pi / N)
            w2 = np.sin(np.arange(0, N) * 2 * np.pi / N)
            i = N
            while i <= n:
                a1 = 2 * (sum(y[i - N:i] * w1)) / N
```

```
b1 = 2 * (sum(y[i - N:i] * w2)) / N

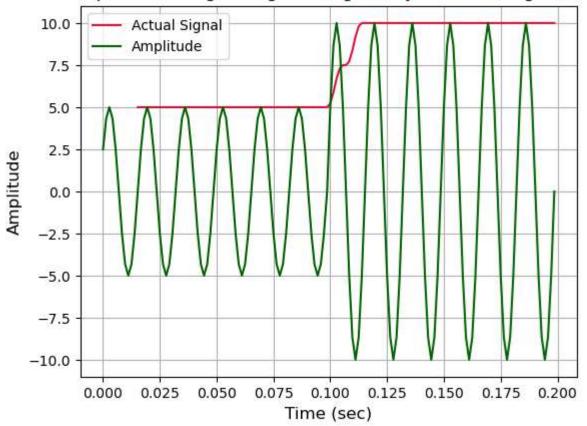
amplitude = np.append(amplitude, np.sqrt(a1 ** 2 + b1 ** 2))
angle = np.append(angle, np.degrees(np.arctan(b1 / a1)))

i = i + 1

return amplitude, angle, time
```

Applying Fourier algorithm on input signal:

Amplitude of Orignial Signal using Full-Cycle Fourier Algorithm

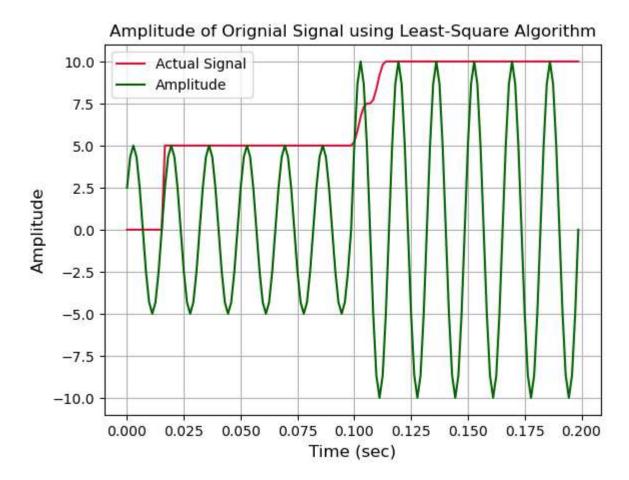


b) Estimate the amplitude of the signal using Least Square algorithm

Least Square algorithm function:

```
In [6]: # LEAST SQUARE ALGORITHM
        def LeastSquareAlgorithm(signal, frequency, sample):
            y = signal
            amplitude = np.zeros_like(y)
            angle = np.zeros like(y)
            f = frequency
            N = sample
            dT = 1/(f*N)
            w0 = 2 * np.pi * f
            A = np.zeros((N, 4))
            t = np.arange(0, 0.2, dT)
            # CALCULATING A MATRIX
            for j in range(1, N+1):
                MF = j - (N + 1) / 2 # Multiplying factor
                A[j - 1, :] = [np.sin(w0 * dT * MF), np.cos(w0 * dT * MF),
                                np.sin(2 * w0 * dT * MF), np.cos(2 * w0 * dT * MF)]
            # CALCULATING PHASOR
            for i in range(N, len(y)):
                B = y[i - N + 1:i + 1]
                X = np.linalg.lstsq(A, B, rcond=None)[0]
                amplitude[i] = np.sqrt(X[0] ** 2 + X[1] ** 2)
                angle[i] = np.arctan2(X[1], X[0])
            return amplitude, angle, t
```

Applying Least Square algorithm on the input signal:



c) Add second harmonic to the signal and repeat a and b (in this case Least Square algorithm considers second harmonics).

```
\left\{ egin{array}{l} 5 sin(\omega t + \pi/6) + 3 sin(2\omega t + \pi/6) \ \ t < 0.1 \ 10 sin(\omega t + \pi/6) + 3 sin(2\omega t + \pi/6) \ \ \ t \geq 0.1 \end{array} 
ight.
```

Given that, f = 60 Hz, and sampling frequency is 12 samples per cycle.

Plotting the input signal (with harmonics):

```
In [8]: ## PLOT
    time = np.arange(start_time,end_time,dT)
    voltage2 = []

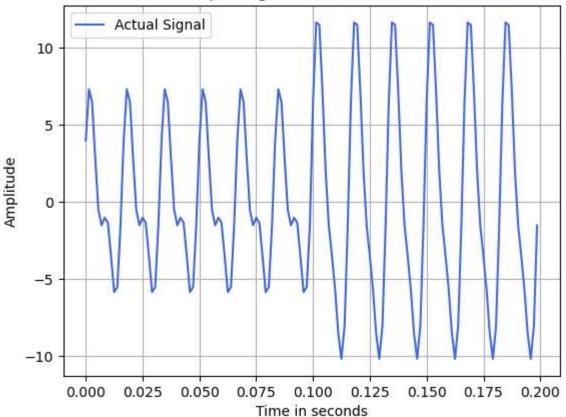
for t in range(0,len(time)):
    if time[t] < 0.1:
        v = (A1 * math.sin(2 * np.pi * f * time[t] + np.pi/6))
        + (A3 * math.sin(2 * 2 * np.pi * f * time[t] + np.pi/6)) # First signal
    else:
        v = A2 * math.sin(2 * np.pi * f * time[t] + np.pi/6)
        + (A3 * math.sin(2 * 2 * np.pi * f * time[t] + np.pi/6)) # Second signal
    voltage2.append(v)

voltage2 = np.array(voltage2)

plt.plot(time,voltage2,mcolors.CSS4_COLORS['royalblue'])
plt.title("Input Signal with Harmonics")</pre>
```

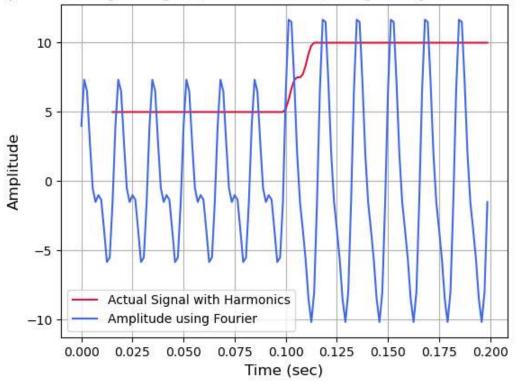
```
plt.xlabel("Time in seconds")
plt.ylabel("Amplitude")
plt.legend(["Actual Signal"])
plt.grid()
plt.show()
```





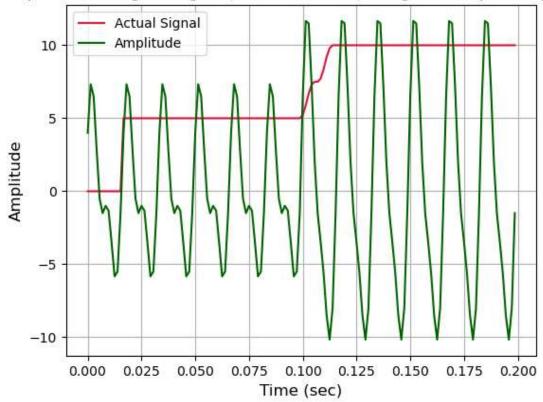
Applying Fourier algorithm on the input signal (with harmonics):

Amplitude of Orignial Signal (with Harmonics) using Full-Cycle Fourier Algorithm



Applying Least Square algorithm on the input signal:

Amplitude of Orignial Signal (with Harmonics) using Least-Square Algorithm



d) Draw frequency response of Cosine filter of Full-cycle Fourier Algorithm with 12 samples per cycle

```
In [11]: coefficient = np.arange(-(N-1)/2,(N+1)/2)
         w0 = 2 * np.pi * f
         y = 2 * np.cos(coefficient * w0 * dT)/N # type of filter function
         w = np.arange(0, 8*w0*dT, 0.1*w0*dT)
         # Z-TRANSFORM COEFFICIENT
         Z = np.zeros((w.shape[0],y.shape[0]),dtype=complex)
         for i in w:
             Z[j,:].real = np.cos(coefficient * i)
             Z[j,:].imag = np.sin(coefficient * i)
             j=j+1
         # VALUE OF FILTER TRANSFER FUNCTION
         H = np.matmul(Z,y)
         w_normalized = (w / (w0*dT))
         # PLOTTING
         plt.plot(w_normalized,np.abs(H))
          plt.xlabel('Normalized frequency (w/w0)')
          plt.ylabel('Amplitude(Output/Input)')
         plt.title('Frequency Response of the Cosine Filter')
          plt.grid()
         plt.show()
```

