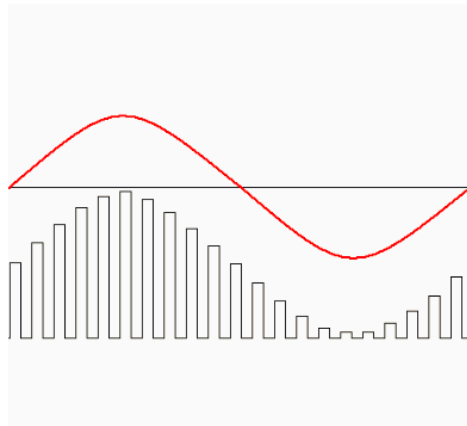


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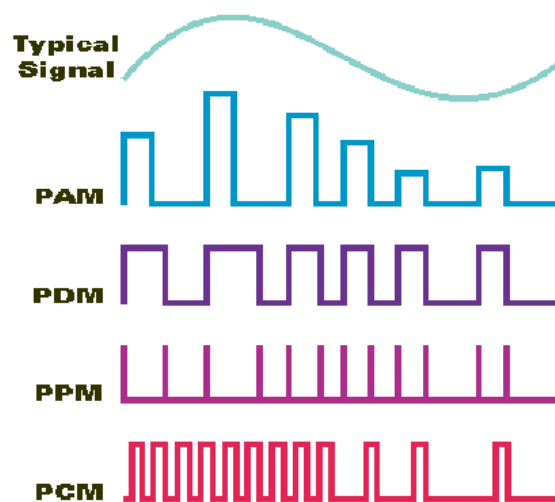
Question No. - 6

PAM



PPM

Modulation



CODE:

(a) Compute the waveform of the PAM signal $s(t)$:

```

import numpy as np
import matplotlib.pyplot as plt

def flat_top_pam_signal(fs, f0, T, flat_top_duration, duration):
    t = np.arange(0, duration, 1/fs)
    signal = np.sin(2 * np.pi * f0 * t)

    samples_per_period = int(fs * T)
    flat_top_samples = int(fs * flat_top_duration)

    pam_signal = np.zeros_like(t)

    for i in range(0, len(t), samples_per_period):
        flat_top_start = i + int((samples_per_period - flat_top_samples) / 2)
        flat_top_end = flat_top_start + flat_top_samples
        pam_signal[flat_top_start:flat_top_end] = signal[i]

    for i in range(len(pam_signal)):
        if (pam_signal[i] < 0 and i<140):
            pam_signal[i] = 0

        if (pam_signal[i] < 0 and (i>280 and i <420)):
            pam_signal[i] = 0

        if (pam_signal[i] > 0 and (i>140 and i <280)):
            pam_signal[i] = 0
        if (pam_signal[i] > 0 and i>420 ):
            pam_signal[i] = 0

```

```

        if (pam_signal[i] > 0 and (i>140 and i <280)):
            pam_signal[i] = 0
        if (pam_signal[i] > 0 and i>420 ):
            pam_signal[i] = 0

    return t, pam_signal

# Parameters
fs = 8000 # Sampling frequency in Hz
f0 = 10000 / (2 * np.pi) # Frequency of the sinusoidal signal in Hz
T = 500e-6 # Pulse duration in seconds
flat_top_duration = T # Duration of flat-top sampling
duration = 0.07 # Total duration of the signal in seconds

# Generate flat-top PAM signal
t_pam, pam_signal = flat_top_pam_signal(fs, f0, T, flat_top_duration, duration)

# Plot the flat-top PAM signal
plt.figure(figsize=(10, 5))
plt.plot(t_pam, pam_signal, label='Flat-Top PAM Signal', color='orange')
plt.xlabel('Time (sec)')
plt.ylabel('Amplitude')
plt.title('Flat-Top PAM Signal with Sample-and-Hold')
plt.legend()
plt.grid(True)
plt.show()

```

(b) Compute $|S(f)|$, denoting the magnitude spectrum of the PAM signal $s(t)$:

```
from scipy.fft import fft

# Compute the FFT of the PAM signal
N = len(t_pam)
frequencies = np.fft.fftfreq(N, 1/fs)
spectrum = fft(pam_signal)
magnitude_spectrum = np.abs(spectrum)

# Plot the magnitude spectrum
plt.figure(figsize=(10, 5))
plt.plot(frequencies, magnitude_spectrum, label='|S(f)|')
plt.xlabel('Frequency (Hz)')
plt.ylabel('|S(f)|')
plt.title('Magnitude Spectrum of the Flat-Top PAM Signal')
plt.legend()
plt.grid(True)
plt.show()
```

(c) Compute the envelope of $|S(f)|$ and confirm the frequency at which it goes through zero for the first time:

```

from scipy.signal import hilbert

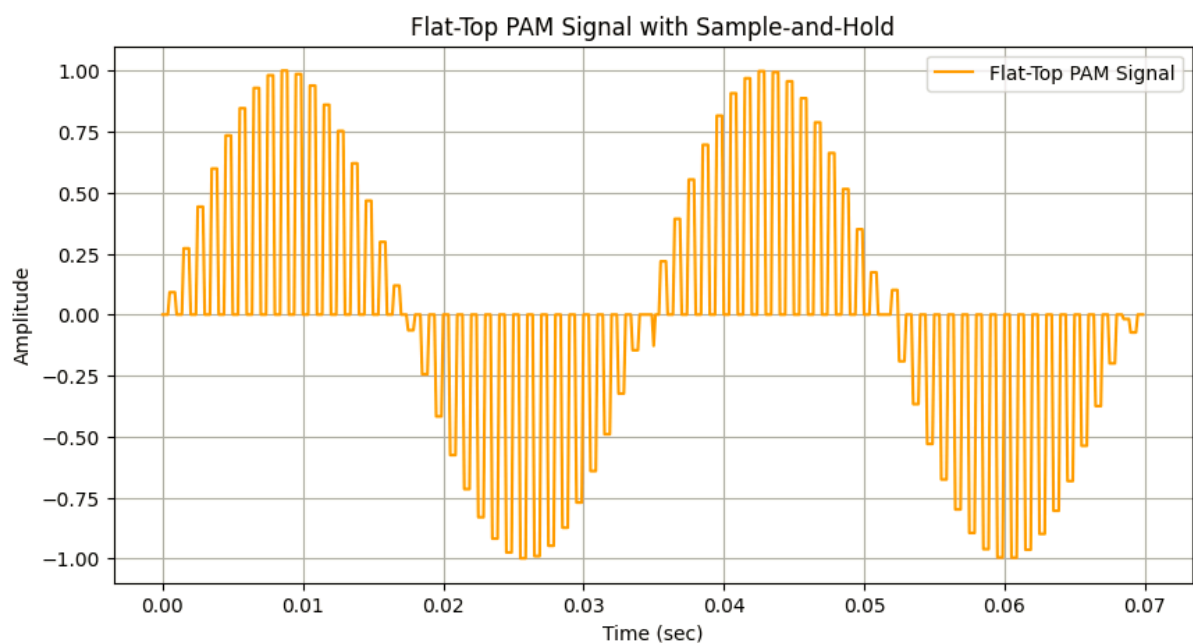
# Compute the envelope of |S(f)|
analytic_signal = hilbert(np.real(spectrum))
envelope = np.abs(analytic_signal)

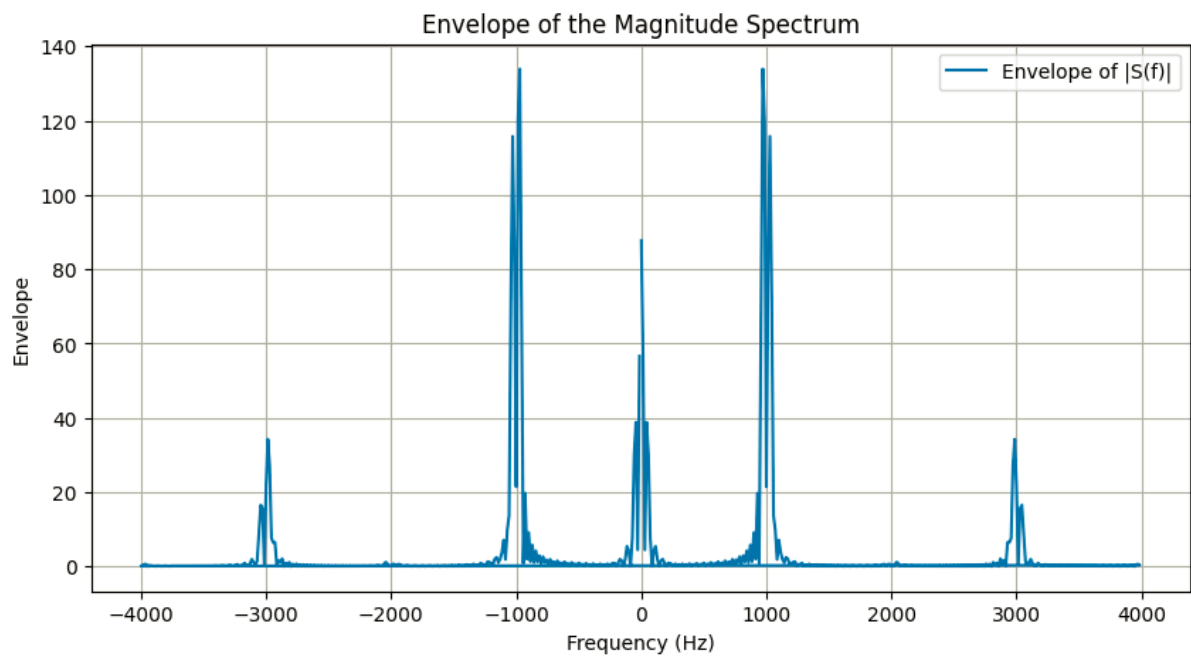
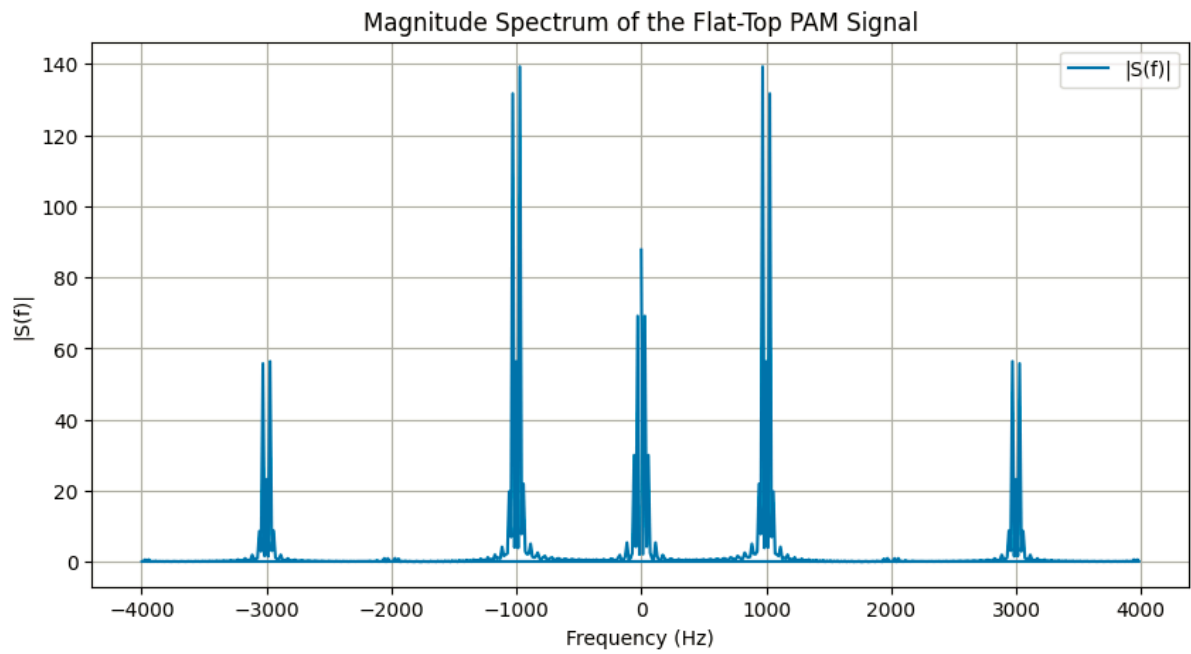
# Plot the envelope
plt.figure(figsize=(10, 5))
plt.plot(frequencies, envelope, label='Envelope of |S(f)|')
plt.xlabel('Frequency (Hz)')
plt.ylabel('Envelope')
plt.title('Envelope of the Magnitude Spectrum')
plt.legend()
plt.grid(True)
plt.show()

# Find the frequency at which the envelope goes through zero for the first time
zero_crossings = np.where(np.diff(np.sign(envelope)))[0]
if len(zero_crossings) > 0:
    first_zero_frequency = frequencies[zero_crossings[0]]
    print(f'The frequency at which the envelope goes through zero for the first time: {first_zero_frequency} Hz')
else:
    print('No zero-crossings found in the envelope.')

```

Graphs:





Question No. - 7

Code :

```
import numpy as np
import matplotlib.pyplot as plt

def continuous_signal(t, f0):
    return np.cos(2 * np.pi * f0 * t)

def undersampled_signal(n, f0, fs):
    Ts = 1 / fs
    sampled_values = continuous_signal(n * Ts, f0)
    return sampled_values

# Signal parameters
f0 = 1 # Frequency of the original signal
fs = 1.5 * f0 # Sampling rate

# Time vector for continuous signal
t_continuous = np.linspace(0, 4, 1000, endpoint=False)

# Time vector for undersampled signal
n_undersampled = np.arange(0, 4, 1 / fs)

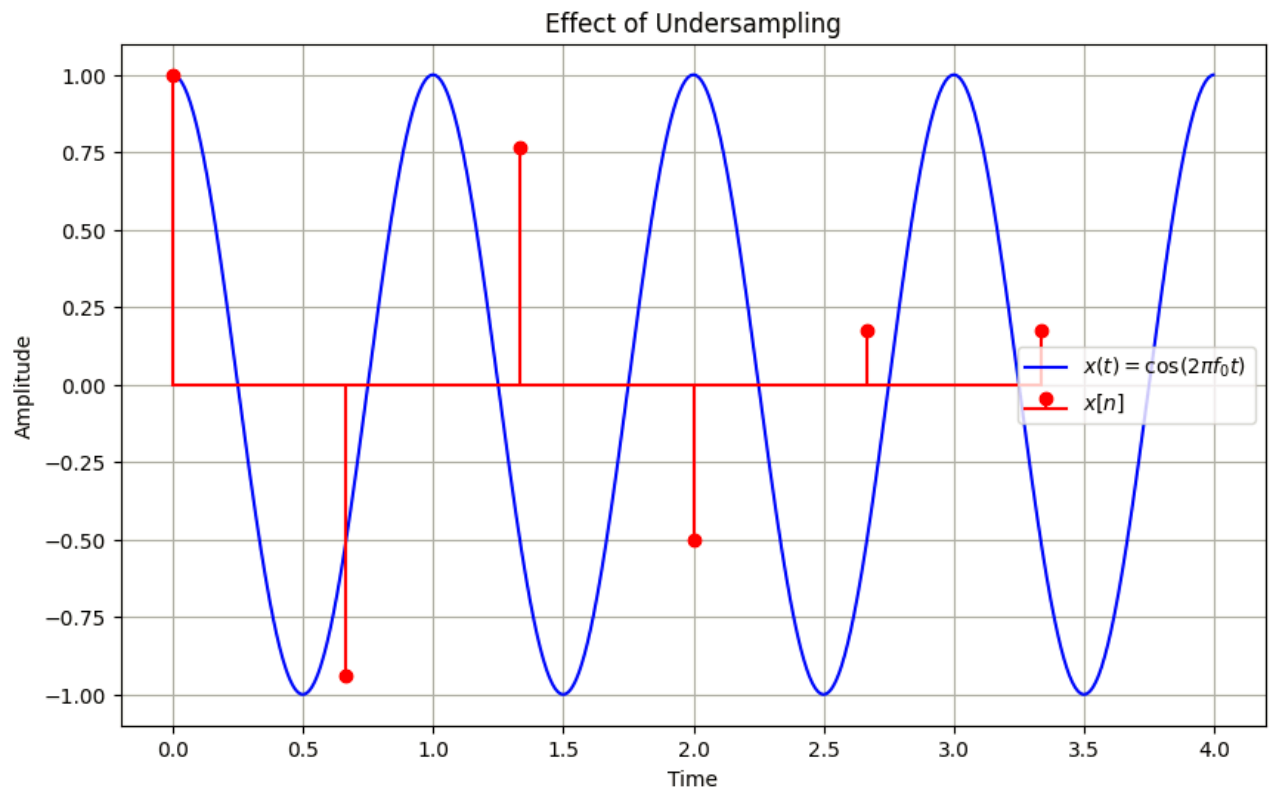
# Compute continuous and undersampled signals
x_continuous = continuous_signal(t_continuous, f0)
x_undersampled = undersampled_signal(n_undersampled, f0, fs)

# Plotting
plt.figure(figsize=(10, 6))

# Plot continuous signal
plt.plot(t_continuous, x_continuous, label=r'$x(t) = \cos(2\pi f_0 t)$', color='blue')

# Plot undersampled signal
plt.stem(n_undersampled, x_undersampled, label=r'$x[n]$', basefmt='r-', linefmt='r-', markerfmt='ro')

plt.title('Effect of Undersampling')
plt.xlabel('Time')
plt.ylabel('Amplitude')
plt.legend()
plt.grid(True)
plt.show()
```



Theory :

$$x(t) = \cos(2\pi f_0 t)$$

The Nyquist rate is defined as twice the maximum frequency component in the signal, so in this case, it would be $2f_0$

Given that the sampling rate f_s is 1.5 times the signal frequency f_0 , the undersampling factor can be calculated as $f_s/(2f_0) = 1.5/2 = 0.75$

convolution to simulate the effect of undersampling.

The sampled signal $x[n]$ can be obtained by convolving the continuous signal $x(t)$ with a train of impulses representing the sampling process.

$$x[n] = x(t) * \sum \delta(t - kT_s)$$

where $T_s = 1/f_s$ is the sampling interval, and $\delta(t)$ is the Dirac delta function.

$$x[n] = \sum x(kT_s) \cdot \delta[n - k]$$

Assuming $f_0 = 1$, the continuous signal $x(t)$ is $\cos(2\pi t)$, and the undersampling factor is 0.75 (undersampling).

$$x[n] = \sum \cos(2\pi k \cdot 0.75) \cdot \delta[n - k]$$