

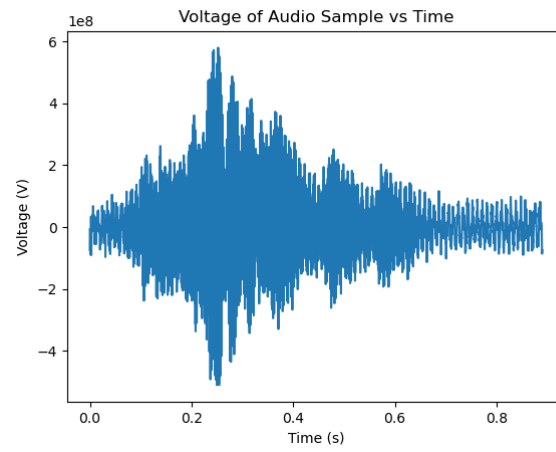
MCEN - 3047

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0.1 Problem 1

0.1.1 Part A



0.1.2 Part B

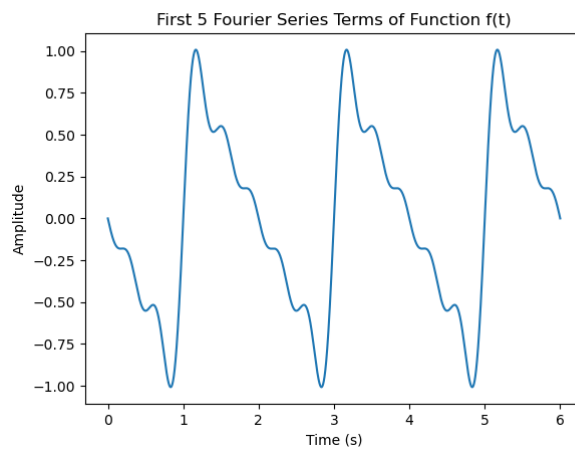
$$\text{LSB} = 8\text{-Bit}$$

0.1.3 Part C

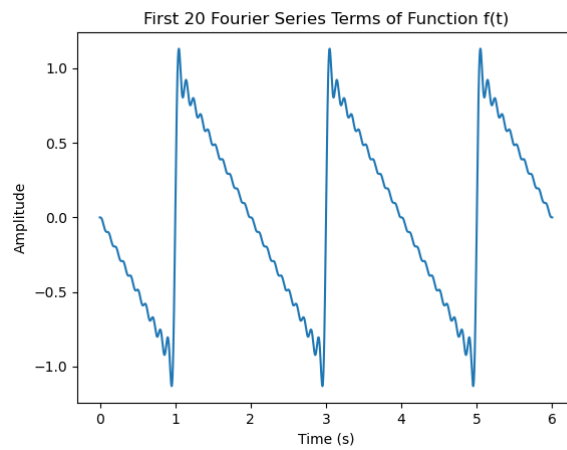
$$V_{fs} = 709859328.0$$

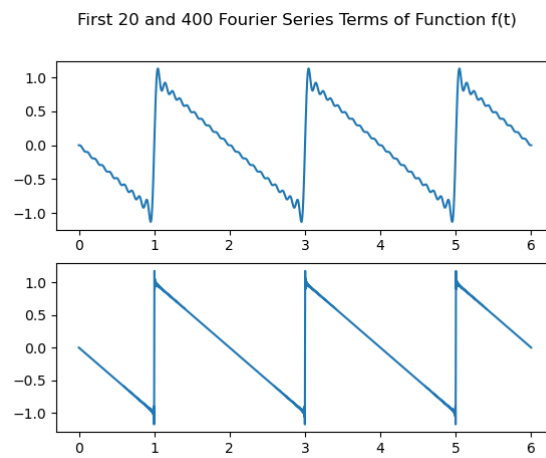
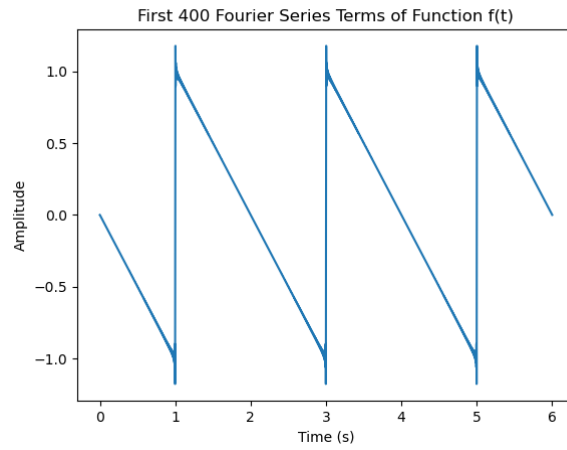
0.2 Problem 2

0.2.1 Part A



0.2.2 Part B



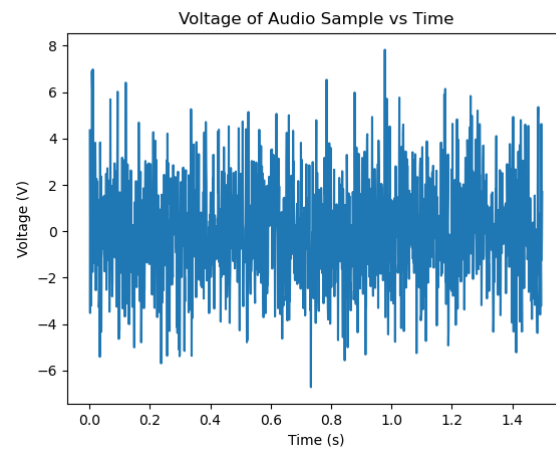


0.2.3 Part C

- The function becomes a better approximation of the cyclical ramp function as n increases, The function appears to become more jagged as n increases as well, better approximating digital signals, if it were one.

0.3 Problem 3

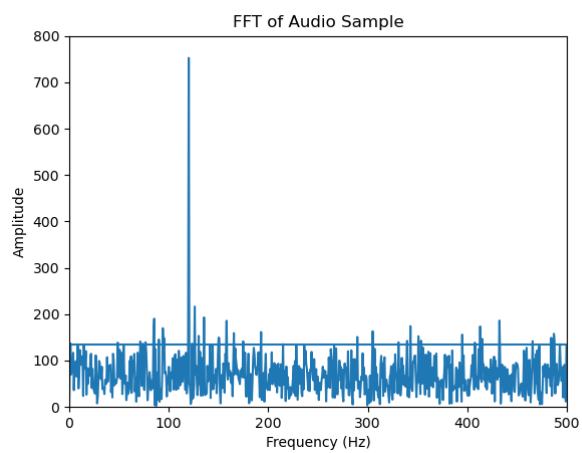
0.3.1 Part A



0.3.2 Part B

It may be possible to get an approximation from this time-series data, however the number may be far from accurate. Thus, one cannot decipher frequency information.

0.3.3 Part C



0.3.4 Part D

Frequency: 120.0800533689126

0.4 CODE

```
import sounddevice as sd
import scipy.io.wavfile as wav
import numpy as np
import matplotlib.pyplot as plt
import torch as tc
import pandas as pd
import os

# I Like Classes
class Set1:
    class Problem1:
        # Calculates the Voltage Range of the DAQ and the
        # ↳ Least Significant Bit
        def get_range(self, recording):
            return self.get_step(recording) * (2 ** np.
                ↳ ceil(self.get_nlsb(recording)))

        # finds the smallest difference between two samples
        @staticmethod
        def get_step(recording):
            for i in range(recording.shape[0] - 1):
                diff = np.abs(recording[i] -
                    ↳ recording[i+1])

            return np.min(diff)

        # Calculates the bit number of least significant
        # ↳ bit
        def get_nlsb(self, recording):
            v_max = np.max(recording)
            # print(v_max)
            return np.log2(v_max / self.get_step(
                ↳ recording))

        def record_audio(duration=7, fs=44100):
            recording = sd.rec(int(duration * fs),
                ↳ samplerate=fs, channels=2)
            sd.wait()
            return recording
```

```

@staticmethod
def get_num_samples( fs=44100, duration=7):
    return int(fs * duration)

def get_time_vector(self, fs=44100, duration=7):
    num_samples = self.get_num_samples(fs=fs,
    ↪ duration=duration)
    # print(num_samples)
    # print("test")
    return np.linspace(start=0, stop=duration,
    ↪ num=num_samples)

def plot_audio_voltage(self, recording, fs=44100):
    time = self.get_time_vector(fs=fs, duration=
    ↪ len(recording)/fs)
    fig = plt.figure()
    plt.title('Voltage of Audio Sample vs Time')
    plt.xlabel('Time (s)')
    plt.ylabel('Voltage (V)')
    plt.plot(time, recording)
    plt.show()

@staticmethod
def plot_audio_fft(recording, fs=44100):

    # recording = recording / np.max(np.abs(
    ↪ recording)) # Normalize amplitude
    # db_data = 20 * np.log10(np.abs(recording)
    ↪ + 1e-10)

    fig = plt.figure()
    plt.title('FFT of Audio Sample')
    plt.xlim(0, 7000)
    plt.ylim(0, 10e11)
    plt.xlabel('Frequency (Hz)')
    plt.ylabel('Amplitude')
    plt.plot(np.fft.fftfreq(len(recording), 1/fs
    ↪ ), np.abs(np.fft.fft(recording)))
    plt.show()

def run(self, cat=False):

    if cat:
        sample_rate, recording = wav.read('
        ↪ ../../data/freyja.wav', 'rb')

```



```

        self.plot_audio_voltage(recording=
            ↳ recording, fs=sample_rate)
        self.plot_audio_fft(recording, fs=
            ↳ sample_rate)
    else:
        recording = self.record_audio()
        self.plot_audio_voltage(recording)
        self.plot_audio_fft(recording)

    lsb = self.get_nlsb(recording)
    print(f"LSB: {np.ceil(lsb)}")
    V_fs = self.get_range(recording)
    print(f"V_fs: {V_fs}")

class Problem2:
    @staticmethod
    def compute_function(n=5,duration=6, fs=44100):
        t = np.linspace(start=0, stop=duration, num=
            ↳ int(fs*duration))
        ten = [(2 * np.cos(np.pi * (i+1)) * np.sin(
            ↳ np.pi * (i+1) * t )) / (np.pi * (i+1)
            ↳ ) for i in range(n)]
        ten = tc.tensor(np.array(ten))
        return ten.sum(dim=0), t

    def plot_n_fourier_terms(self, nt=5, duration=6, fs
        ↳ =44100):
        ft, t = self.compute_function(n=nt, duration
            ↳ =duration, fs=fs)
        fig = plt.figure()
        plt.title(f'First {nt} Fourier Series Terms
            ↳ of Function f(t)')
        plt.xlabel('Time (s)')
        plt.ylabel('Amplitude')
        plt.plot(t,ft)
        plt.show()

    def double_plot_n_fourier_terms(self, n1=20, n2
        ↳ =400, duration=6, fs=44100):
        f1, t = self.compute_function(n=n1, duration
            ↳ =duration, fs=fs)
        f2, t = self.compute_function(n=n2, duration
            ↳ =duration, fs=fs)

```

```

fig, axs = plt.subplots(2)
fig.suptitle('First 20 and 400 Fourier
    ↳ Series Terms of Function f(t)')
axs[0].plot(t, f1)
axs[1].plot(t, f2)
plt.show()

def run(self):
    self.plot_n_fourier_terms(nt=5)
    self.plot_n_fourier_terms(nt=20)
    self.plot_n_fourier_terms(nt=400)
    self.double_plot_n_fourier_terms()

class Problem3:
    """ 1. Takes a csv file, tone, that contains audio
        ↳ data.
        i. The first column is the time vector
        ii. The second column is the audio data
        2. Executes a FFT on the audio
        3. plots the timeseries data
        4. plots the FFT data
        5. Saves the plots as images
        6. calculates the frequency of the tone from the
            ↳ least significant bit of the DAQ
            and voltage scale"""
    def run(self):
        df = self.read_csv()
        self.plot_audio(df)
        self.plot_audio_fft(df)
        self.calculate_peak_frequency(df)
        # self.save_plots(df)

    def read_csv(self, file_path='tone.csv', home=True)
        ↳ :
        if home:
            file_path = os.path.join('../..data
                ↳ /', file_path)

            df = pd.read_csv(file_path)

            df.columns = ['time', 'audio']
        return df

```

```

def plot_audio(self, df):
    fig = plt.figure()
    plt.title('Voltage of Audio Sample vs Time')
    plt.xlabel('Time (s)')
    plt.ylabel('Voltage (V)')
    plt.plot(df['time'], df['audio'])
    plt.show()

def plot_audio_fft(self, df):

    sample_rate = (df['time'][1] - df['time']
        ↪ ) [0]) ** -1
    fig = plt.figure()
    plt.title('FFT of Audio Sample')
    plt.xlim(0, 500)
    plt.ylim(0, 800)
    plt.xlabel('Frequency (Hz)')
    plt.ylabel('Amplitude')
    plt.plot(np.fft.fftfreq(len(df['audio']), 1/
        ↪ sample_rate), np.abs(np.fft.fft(df['
        ↪ audio'])))
    plt.show()
    def save_plots(self, df, file_path='results'
        ↪ , home=True):
        if home:
            file_path = os.path.join('../..', file_path
                ↪ )

        self.plot_audio(df)
        self.plot_audio_fft(df)
        print("Saving plots...")
        plt.savefig(file_path + '/problem3.png')

def calculate_peak_frequency(self, df):
    sample_rate = (df['time'][1] - df['time']
        ↪ ) [0]) ** -1
    freq_bins = np.fft.fftfreq(len(df['audio']),
        ↪ 1/sample_rate)
    freq_amps = np.abs(np.fft.fft(df['audio']))

    loc, val = np.argmax(freq_amps), np.max(
        ↪ freq_amps)
    print(f"Peak Frequency: {freq_bins[loc]}")

def run(self):

```

```
        """ Runs the Entire problem set """

        print("Running_Problem_1")
        p1 = self.Problem1()
        p1.run(cat=True)

        print("Running_Problem_2")
        p2 = self.Problem2()
        p2.run()

        print("Running_Problem_3")
        p3 = self.Problem3()
        p3.run()

# Easy Execution with "python set_1.py"
s1 = Set1()
print("Running_Set_1")
s1.run()
```