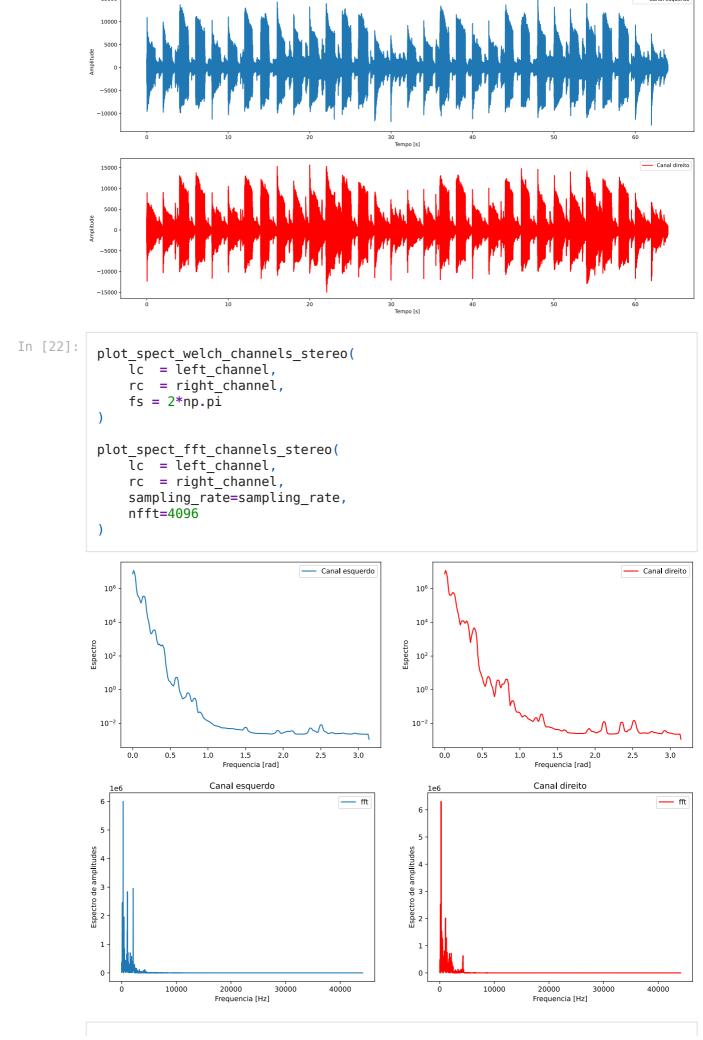
```
In [16]:
          # -*- coding: utf-8 -*-
          import numpy as np
          import matplotlib.pyplot as plt
          from scipy.io import wavfile
          from scipy.signal import welch, lfilter
          from scipy import fftpack
In [17]:
          def plot time domain channels stereo(time, lc, rc):
              #Plota as figuras ao longo do tempo
              #Plota os canais esquerdo e direito
              plt.figure(1, figsize=(20, 5))
              plt.plot(time, lc, label="Canal esquerdo")
              plt.legend()
              plt.xlabel("Tempo [s]")
              plt.ylabel("Amplitude")
              plt.show()
              plt.figure(2,figsize=(20, 5))
              plt.plot(time, rc, color="red", label="Canal direito")
              plt.legend()
              plt.xlabel("Tempo [s]")
              plt.ylabel("Amplitude")
              plt.show()
In [18]:
          def plot spect welch channels stereo(lc, rc, fs):
              #Sample Frequencies, Power Spectral Density
              sf lc, psd lc = welch(
                  x=lc,
                  fs=fs,
                  window='flattop',
                  nperseg=512,
                  scaling='spectrum'
              sf rc, psd rc = welch(
                  x=rc,
                  fs=fs,
                  window='flattop',
                  nperseg=512,
                  scaling='spectrum'
              )
              #Plota o espectro do sinal para frequencias normalizadas entre 0 1 pi
              #(frequencias positivas)
              plt.subplots(figsize=(15,5))
              plt.subplot(1, 2, 1)
              plt.semilogy(sf lc, psd lc, label="Canal esquerdo")
              plt.legend()
              plt.xlabel('Frequencia [rad]')
              plt.ylabel('Espectro')
              plt.subplot(1, 2, 2)
              plt.semilogy(sf_rc, psd_rc, color="red", label="Canal direito")
              plt.legend()
              plt.xlabel('Frequencia [rad]')
              plt.ylabel('Espectro')
              plt.show()
In [19]:
          def plot_spect_fft_channels_stereo(lc,rc, sampling_rate, nfft):
```

freq\_lc = np.linspace(0., sampling\_rate, nfft) #Interpola para determinal

```
sig fft lc = fftpack.rfft(lc,nfft)
plt.subplots(figsize=(15,5))
plt.subplot(1, 2, 1)
plt.title("Canal esquerdo")
plt.plot(freq lc, np.abs(sig fft lc), label="fft")
plt.legend()
plt.xlabel('Frequencia [Hz]')
plt.ylabel('Espectro de amplitudes')
#plt.plot(freq lc, np.abs(fftpack.fftshift(sig fft lc)), label="fftshift"
plt.legend()
freq rc = np.linspace(0., sampling rate, nfft) #Interpola para determinal
sig fft rc = fftpack.rfft(rc,nfft)
plt.subplot(1, 2, 2)
plt.title("Canal direito")
plt.plot(freg rc, np.abs(sig fft rc), color="red", label="fft")
plt.legend()
plt.xlabel('Frequencia [Hz]')
plt.ylabel('Espectro de amplitudes')
#plt.plot(freq rc, np.abs(fftpack.fftshift(sig fft rc)), color="green",
plt.legend()
plt.show()
```

```
In [20]:
          #Carrega o arquivo
          sampling rate, data = wavfile.read('569127 josefpres dark-loops-201-simple-
          #sampling rate, data = wavfile.read('581010 xcreenplay smoking-in-the-ange
          number of samples = data.shape[0]
          number of channels = data.shape[1]
          #Tempo total = numero de amostras / fs
          duration = number of samples / sampling rate
          #Carrega o arquivo em dois canais (audio estereo)
          left channel = data[:, 0]
          right channel = data[:, 1]
          print(f"Numero de canais = {number of channels}")
          print(f"Duracao = {duration}s")
          print(f'Numero de amostras: {number of samples}')
          print(f"Amostras por segundo: {sampling rate}Hz")
         Numero de canais = 2
         Duracao = 64.0s
         Numero de amostras: 2822400
         Amostras por segundo: 44100Hz
         /tmp/ipykernel 342464/1384073729.py:2: WavFileWarning: Chunk (non-data) not u
         nderstood, skipping it.
           sampling rate, data = wavfile.read('569127 josefpres dark-loops-201-simpl
         e-mix-2-short-loop-60-bpm.wav')
In [21]:
          #Interpola para determinar eixo do tempo
          time = np.linspace(0., duration, number of samples)
          plot time domain channels stereo(
              time=time,
              lc=left channel,
              rc=right channel
          )
```



```
#Dizimando o sinal pelo fator M
In [23]:
              M=4
              decimated_lc = left_channel[0:-1:M]
              decimated_rc = right_channel[0:-1:M]
In [24]:
              plot_spect_welch_channels_stereo(
                    lc = decimated_lc,
                   rc = decimated_rc,
                   fs = (2*np.pi)/M
              plot_spect_fft_channels_stereo(
                   lc = decimated_lc,
rc = decimated_rc,
                    sampling_rate=sampling_rate//M,
                   nfft=4096
              )
                                                      Canal esquerdo
                                                                                                             Canal direito
              10<sup>4</sup>
                                                                       104
              10<sup>2</sup>
                                                                       102
              10<sup>0</sup>
                                                                       10º
                                                         0.7
                                                               0.8
                   0.0
                        0.1
                              0.2
                                                    0.6
                                                                            0.0
                                                                                 0.1
                                                                                      0.2
                                                                                                            0.6
                                                                                                                       0.8
                                                                                                 0.4
                                                                                                                  0.7
                                                                                             Canal direito
                                   Canal esquerdo
              4.0
                                                                                                                     — fft
              3.5
              3.0
             Espectro de amplitudes
0. 1. 2. 0
0. 2. 1. 0
                                                                       Espectro de amplitudes
                                                                         2
              0.5
              0.0
                                  4000
                                         6000
                                                 8000
                                                         10000
                                                                                   2000
                                                                                                  6000
                                                                                                          8000
                                                                                                                  10000
                          2000
                                                                                           4000
                                    Frequencia [Hz]
                                                                                             Frequencia [Hz]
In [25]:
              #Carrega os coeficientes do filtro
              b = np.genfromtxt('coeffs.csv', delimiter=',')
              #Plota coeficientes do filtro FIR
              plt.figure(7, figsize=(20, 5))
              plt.stem(b)
              plt.show()
```

10<sup>7</sup>

0.5

```
0.25 -

0.20 -

0.15 -

0.00 -

0.00 -

0.00 -

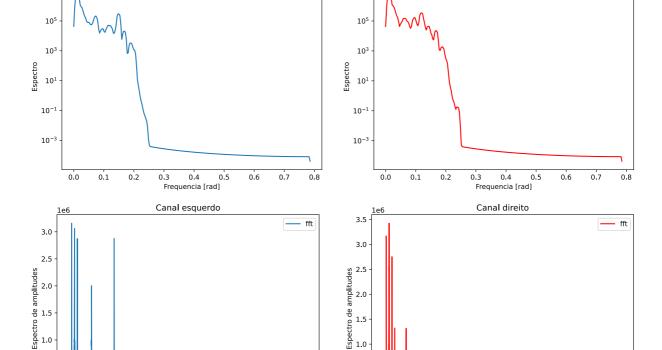
0 20 40 60 80 100
```

```
In [26]: #Filtra os dados dos canais esquerdo e direito
    decimated_filtered_lc = lfilter(b, 1, decimated_lc)
    decimated_filtered_rc = lfilter(b, 1, decimated_rc)
```

```
In [27]:
    plot_spect_welch_channels_stereo(
        lc = decimated_filtered_lc,
        rc = decimated_filtered_rc,
        fs = (2*np.pi)/M
)

plot_spect_fft_channels_stereo(
        lc = decimated_filtered_lc,
        rc = decimated_filtered_rc,
        sampling_rate=sampling_rate//M,
        nfft=4096
)
```

Canal esquerdo



0.5

107

```
In [28]: # Escrita de arquivo dizimado
    audio = np.array([decimated_lc, decimated_rc]).T
    scaled = np.int16(audio/np.max(np.abs(audio)) * 32767)
    filename = 'signal_decimated_' + str(M) + '.wav'
    wavfile.write(filename, sampling_rate//M, scaled)
```

10000

6000

Frequencia [Hz]

8000

8000

6000

Frequencia [Hz]

10000

Canal direito

```
In [29]:
           # Escrita de arquivo dizimado e filtrado
           audio = np.array([decimated_filtered_lc, decimated_filtered_rc]).T
           scaled = np.int16(audio/np.max(np.abs(audio)) * 32767)
           filename = 'signal decimated_' + str(M) + '_filted.wav'
           wavfile.write(filename, sampling rate//M, scaled)
In [30]:
           # Fator de interpolação
           interpolated lc = np.zeros(L * len(left channel))
           interpolated lc[::L] = left channel
           interpolated rc = np.zeros(L * len(right channel))
           interpolated rc[::L] = right channel
In [31]:
           plot spect welch channels stereo(
                lc = interpolated lc,
                rc = interpolated rc,
                fs = (2*np.pi)*L
           plot_spect_fft_channels_stereo(
                lc = interpolated_lc,
                rc = interpolated rc,
                sampling rate=sampling rate*L,
                nfft=4096
                   Canal esquerdo

    Canal direito

            104
                                                           104
            10
                                                           102
                                                           10<sup>0</sup>
            10
            10-2
                                                           10-2
                              6
Frequencia [rad]
                                                  12
                                                                             6
Frequencia [rad]
                                                                                           10
                                                                              Canal direito
                               Canal esquerdo
            500000
                                                          600000
            400000
                                                          500000
                                                          400000
            300000
                                                          300000
            200000
                                                          200000
            100000
                                                          100000
                     25000
                          50000
                               75000 100000 125000 150000 175000
                                                                   25000
                                                                        50000
                                                                             75000 100000 125000 150000 175000
In [32]:
           #Filtra os dados dos canais esquerdo e direito
           interpolated_filtered_lc = lfilter(b, 1, interpolated_lc)
           interpolated_filtered_rc = lfilter(b, 1, interpolated_rc)
```

```
In [33]:
            plot spect welch channels stereo(
                 lc = interpolated_filtered_lc,
                 rc = interpolated_filtered_rc,
                 fs = (2*np.pi)*L
            plot spect fft channels stereo(
                 lc = interpolated_filtered_lc,
                 rc = interpolated_filtered_rc,
                 sampling_rate=sampling rate*L,
                 nfft=4096
            )

    Canal esquerdo

    Canal direito

             10<sup>4</sup>
                                                               104
             10<sup>2</sup>
                                                                10<sup>2</sup>
             10<sup>0</sup>
                                                               10<sup>0</sup>
             10-
                                                               10-2
             10-4
                                                               10-4
                                                10
                                                      12
                                                                                                  10
                                                                                                        12
                                 Frequencia [rad]
                                                                                   Frequencia [rad]
                                 Canal esquerdo
                                                                                   Canal direito
                                                              600000
             500000
                                                              500000
             400000
                                                              400000
             300000
             200000
                                                              200000
             100000
                                                              100000
                0
                       25000
                            50000
                                  75000 100000 125000 150000 175000
                                                                        25000
                                                                             50000
                                                                                   75000 100000 125000 150000 175000
                                  Frequencia [Hz]
                                                                                   Frequencia [Hz]
In [34]:
            # Escrita de arquivo interpolado
            audio = np.array([interpolated lc, interpolated rc]).T
            scaled = np.int16(audio/np.max(np.abs(audio)) * 32767)
            filename = 'signal_interpolated_' + str(L) + '.wav'
            wavfile.write(filename, sampling rate*L, scaled)
In [35]:
            # Escrita de arquivo interpolado e filtrado
            audio = np.array([interpolated_filtered_lc, interpolated_filtered_rc]).T
            scaled = np.int16(audio/np.max(np.abs(audio)) * 32767)
            filename = 'signal_interpolated_' + str(L) + '_filted.wav'
            wavfile.write(filename, sampling rate*L, scaled)
```