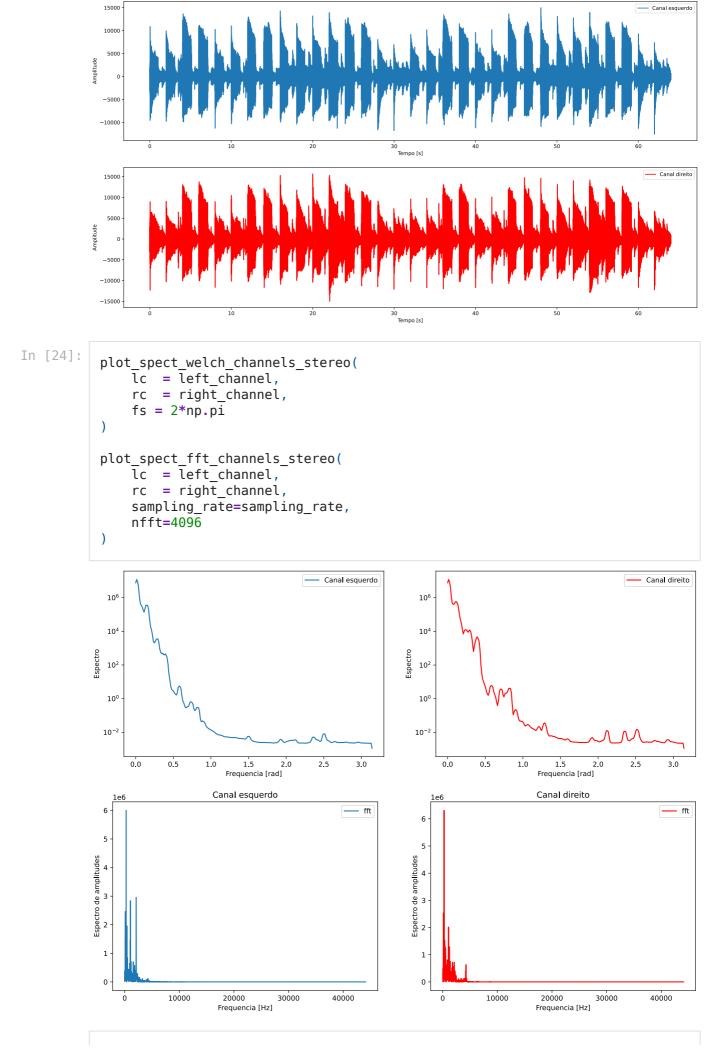
```
In [18]:
          # -*- coding: utf-8 -*-
          import numpy as np
          import matplotlib.pyplot as plt
          from scipy.io import wavfile
          from scipy.signal import welch, lfilter
          from scipy import fftpack
In [19]:
          def plot time domain channels stereo(time, 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 [20]:
          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 [21]:
          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 [22]:
          #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 359008/1384073729.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 [23]:
          #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
          )
```



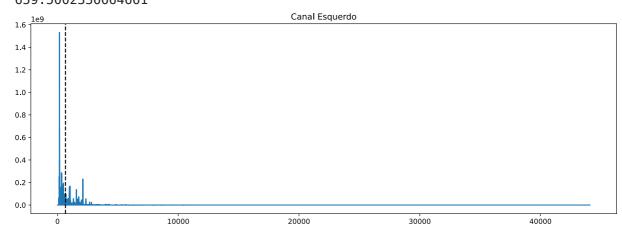
```
#Dizimando o sinal pelo fator M
In [25]:
            decimated_lc = left_channel[0:-1:M]
            decimated rc = right channel[0:-1:M]
In [26]:
            plot_spect_welch_channels_stereo(
                 lc = decimated_lc,
                 rc = decimated_rc,
                 fs = (2*np.pi)/M
            plot spect fft channels stereo(
                 lc = decimated lc,
                 rc = decimated rc,
                 sampling rate=sampling rate//M,
                 nfft=4096
            )

    Canal direito

                                              Canal esquerdo
            10<sup>4</sup>
                                                             10
            10<sup>2</sup>
                                                             102
            10<sup>0</sup>
                                                             10<sup>0</sup>
                                                 0.7
                                                                     0.1
                                                                                                      0.8
                0.0
                    0.1
                         0.2
                                            0.6
                                                                 0.0
                                                                                             0.6
                                                                                                 0.7
                                                                                   0.4
                              Canal esquerdo
                                                                                Canal direito
            4.0
            3.5
            3.0
                                                             Espectro de amplitudes
            2.5
            2.0
           þ
                                                              2
           0
1.5
           1.0
            0.5
            0.0
                                                                       2000
                      2000
                             4000
                                   6000
                                          8000
                                                10000
                                                                             4000
                                                                                    6000
                                                                                           8000
                                                                                                 10000
                               Frequencia [Hz]
                                                                               Frequencia [Hz]
In [27]:
            freq_lc = np.linspace(0., sampling_rate, number_of_samples) #Interpola para d
            sig fft lc = fftpack.rfft(left channel,number of samples)
            plt.figure(1, figsize=(15, 5))
            plt.plot(freq_lc, np.abs(sig_fft_lc[:number_of_samples]))
            #"centro de massa" da função no domínio da frequência
            peso lc = 0.0 #porcentagem do sinal acumulado
            rangeData_lc = len(sig_fft_lc) #frequências positivas da FFT
            sc_lc = sum(abs(sig_fft_lc[:rangeData_lc])) #todo o sinal acumulado
            count_lc = 0 # quantidade de amostras somadas
            for i in abs(sig_fft_lc[:rangeData_lc]):
              peso_lc = peso_lc + i/sc_lc
              count lc = count lc + 1
```

```
if(peso_lc >= 0.5):
    print(peso_lc)
    break
print(freq_lc[count_lc]) #frequência de corte para o primeiro canal
plt.title("Canal Esquerdo")
plt.axvline(x = freq lc[count lc], color = 'black', ls = '--')
plt.show()
freq_rc = np.linspace(0., sampling_rate, number_of_samples) #Interpola para (
sig fft rc = fftpack.rfft(right channel,number of samples)
plt.figure(2, figsize=(15, 5))
plt.plot(freq rc, np.abs(sig fft rc[:number of samples]), color='red')
peso rc = 0.0
rangeData rc = len(sig fft rc)
sc rc = sum(abs(sig fft rc[:rangeData rc]))
count rc = 0
for i in abs(sig_fft_rc[:rangeData_rc]):
 peso_rc = peso_rc + i/sc_rc
  count rc = count rc + 1
  if(peso rc >= 0.5):
    print(peso rc)
    break
print(freq_rc[count_rc])
plt.title("Canal Direito")
plt.axvline(x = freq rc[count rc], color = 'black', ls = '--')
plt.show()
```

0.5000133156517214 659.5002336664661



0.5000178643250714 785.3284032484422

```
Canal Direito

1.4

1.2

1.0

0.6

0.4

0.2

0.0

10000

20000

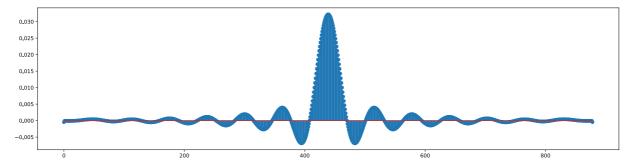
30000

40000
```

```
In [28]:
    aux = np.abs(sig_fft_rc).cumsum()
    meio = aux[-1] / 2
    valorMeio = aux[aux >= meio][0]
    freqMetade = freq_lc[aux >= valorMeio][0]
    print("Frequência de metade do conteúdo espectral: ", freqMetade, " Hz")
```

Frequência de metade do conteúdo espectral: 785.3127782429062 Hz

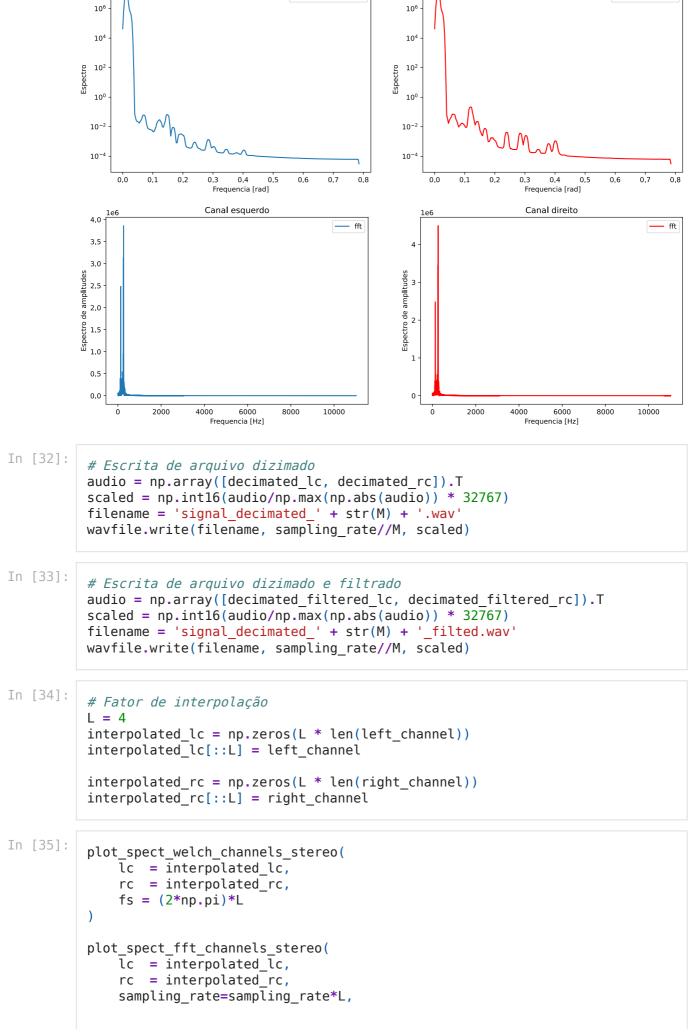
```
In [29]: #Carrega os coeficientes do filtro
b = np.genfromtxt('coeffs_pyfda_lp.csv', delimiter=',')
#Plota coeficientes do filtro FIR
plt.figure(7, figsize=(20, 5))
plt.stem(b)
plt.show()
```

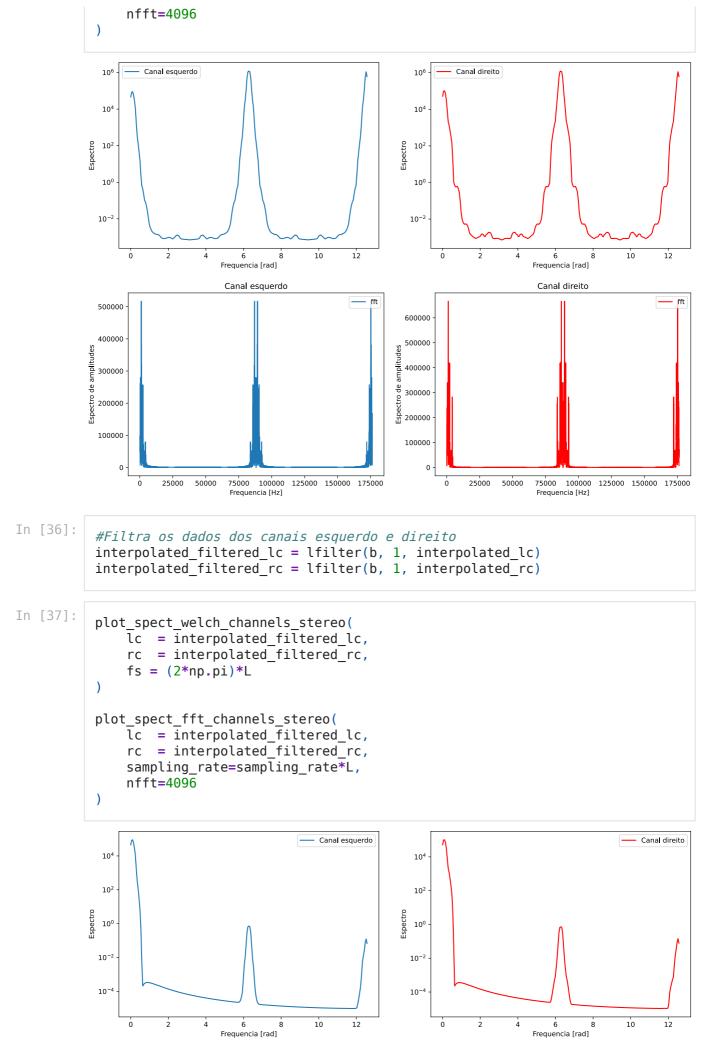


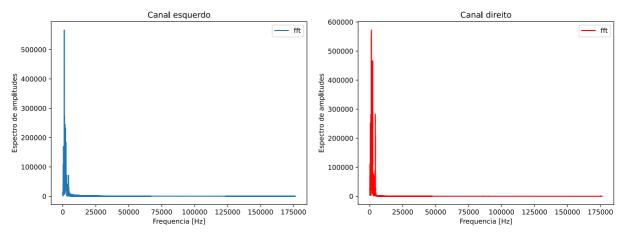
```
In [30]: #Filtra os dados dos canais esquerdo e direito
    decimated_filtered_lc = lfilter(b, 1, decimated_lc)
    decimated_filtered_rc = lfilter(b, 1, decimated_rc)
```

— Canal direito

— Canal esquerdo







```
In [38]: # 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 [39]: # 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)
```