# Calculating frequencies of signal

### **Analysis steps:**

- load signal from file
- get sample rate and array of values
- filter signal with lowpass filter to remove high frequencies
- divide signal into chunks
- for every chunk calculate FFT Fast Fourier Transform
- find maximum value in spectrum
- change it to frequency multiplying it by (fs / chunk\_size)
- calculate MSE for frequency array
- transform MSE into user score

## VoiceAnalyzer class description

### Methods:

- read\_signal reads signal from file
- butter\_lowpass calculate filter params
- butter\_lowpass\_filter filters signal with those params
- divide\_signal\_into\_chunks divides signal into array of chunks
- get\_frequencies calculate frequencies in signal
- trim\_frequencies trims frequencies array to the same length

```
In [19]:
          from scipy.io import wavfile
          import matplotlib.pyplot as plt
          import numpy as np
          from scipy.signal import butter, lfilter, freqz
          from scipy.signal import stft
          from scipy.fft import fft
          class VoiceAnalyzer:
              def read signal(self, path):
                  fs, y = wavfile.read(path)
                  return fs, y
              def butter_lowpass(self, cutoff, fs, order=5):
                  nyq = 0.5 * fs
                  normal cutoff = cutoff / nyq
                  b, a = butter(order, normal cutoff, btype='low', analog=False)
                  return b, a
              def butter_lowpass_filter(self, data, cutoff, fs, order=5):
                  b, a = self.butter_lowpass(cutoff, fs, order=order)
                  y = lfilter(b, a, data)
                  return y
              def divide_signal_into_chunks(self, y, chunk_size):
                  return [y[i: i + chunk_size] for i in range(0, len(y), chunk_size)]
              def get_frequencies(self, y, fs, chunk_size):
                  freqs = []
                  for y_fourier in y:
                      y_fourier = fft(y_fourier)
```

```
y_fourier = y_fourier[0:int(chunk_size / 2)]
        # index of max -> freq, * fs / step to get proper values
       freq = np.argmax(abs(y_fourier)) * (fs / chunk_size)
       # round to 2 decimal places
       freq = round(freq, 2)
       # change freg == 0 to 1, beacause of logarithming later
       if(freq == 0):
            freq = 1
       freqs.append(freq)
    return freqs
def trim_frequencies(self, freq, freq2):
    if len(freq) > len(freq2):
       freq = freq[0:len(freq2)]
   else:
       freq2 = freq2[0:len(freq)]
    return freq, freq2
```

### VoiceScoreCalculator class

#### Methods:

- set\_freq, set\_freq2 frequencies arrays setters
- get\_mse calculatre MSE from 2 signals, uses VoiceAnalyzer class
- get\_frequencies returns frequencies arrays
- process\_mse apply function to MSE, covert big values to small and small to big (quadratic function)  $y = (x 50000 / 1000) ^ 2$
- get\_score function that apply converting methods, also returns 10^6 score for 0 MSE (if any user can achieve such result), and 0 if MSE is bigger than 50 000 (because of poor vocal performance)

```
In [20]:
          class VoiceScoreCalculator:
              def __init__(self):
                  self.CHUNK SIZE = 4410
                  self.LOW_CUT_FREQ = 600
                  self.analyzer = VoiceAnalyzer()
                  self.freq = []
                  self.freq2 = []
              def set freq(self, freq):
                  self.freq = freq
              def set freq2(self, freq2):
                  self.freq2 = freq2
              def get_mse(self, path1, path2):
                  fs, y = self.analyzer.read_signal(path1)
                  fs2, y2 = self.analyzer.read_signal(path2)
                  y = self.analyzer.butter_lowpass_filter(y, self.LOW_CUT_FREQ, fs)
                  y2 = self.analyzer.butter lowpass filter(y2, self.LOW CUT FREQ, fs2)
                  y = self.analyzer.divide_signal_into_chunks(y, self.CHUNK_SIZE)
                  y2 = self.analyzer.divide_signal_into_chunks(y2, self.CHUNK_SIZE)
                  freq = self.analyzer.get_frequencies(y, fs, self.CHUNK_SIZE)
                  freq2 = self.analyzer.get_frequencies(y2, fs2, self.CHUNK_SIZE)
                  freq, freq2 = self.analyzer.trim_frequencies(freq, freq2)
```

```
# set freqs to object properties
    self.set freq(freq)
   self.set_freq2(freq2)
   mse = np.square(np.subtract(freq, freq2)).mean()
# Quadratic function, change small to big, big to small values
def process_mse(self, mse):
   mse = mse - 50000
   mse = mse / 1000
   mse = np.power(mse, 2)
   mse = round(mse)
   return mse
def get frequencies(self):
   return {'freq': self.freq, 'freq2': self.freq2}
def get_score(self, path1, path2):
   mse = self.get_mse(path1, path2)
   # Apply function, else return big number
   if(50000 > mse >= 0):
       return self.process_mse(mse)
   elif(mse >= 50000):
       return 0
   else:
       return np.power(10, 6)
```

### **Example usage**

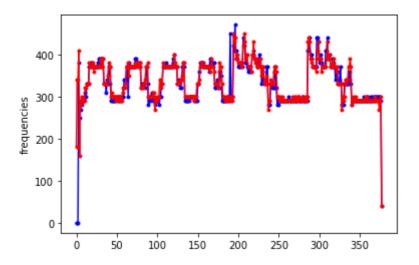
```
In [ ]:
    path1 = 'voice_example2_1.wav'
    path2 = 'voice_example2_2.wav'

    voiceScoreCalculator = VoiceScoreCalculator()
    score = voiceScoreCalculator.get_score(path1, path2)
    freqs = voiceScoreCalculator.get_frequencies()

    freq = freqs['freq']
    freq2 = freqs['freq2']
    score
```

## Plotting result

```
plt.plot(freq, marker=".", color='b')
plt.plot(freq2, marker=".", color='r')
plt.ylabel('frequencies')
plt.show()
```



# Plot MSE processing function

- if MSE is small, user gets high score
- if MSE if big, user gets smaller score

```
In [23]:

def processing_fun(x):
    x = x - 50000
    x = x / 1000
    x = np.power(x, 2)
    return x

x = np.arange(0, 50000, 0.1)
y = processing_fun(x)

plt.figure(2)
plt.plot(x, y)
plt.show()
```

