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In [ ]: # Individual Assignment
        # Mason Wheeler
        # Creating my own 5-band equalizer
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Sub-bass: 20 to 60 Hz Bass: 60 to 250 Hz Lower midrange: 250 to 500 Hz Midrange: 500 to 2 kHz Upper midrange: 2 to 4 kHz Presence: 4 to 6 kHz Brilliance: 6 to 20 kHz

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In [ ]: import numpy as np
        from scipy.io import wavfile
        from scipy import signal
        import simpleaudio as sa

        # Define the cutoff frequencies for the 5 bands in terms of Nyquist rate
        cutoffs = [0.01, 0.05, 0.1, 0.4, 0.8, 1.0] # Adjust as necessary
        orders = [5] * 6 # Use 5th order filters

        # Create filters for the bands
        filters = [signal.butter(order, cutoff, 'low' if i == 0 else 'high' if i == len(cutoffs) - 1,
                                btype='butter', analog=False, output='ba')
                   for i, (order, cutoff) in enumerate(zip(orders, cutoffs))]

        # Function to equalize audio
        def equalize_audio(audio, gains):
            assert len(gains) == len(filters), "Number of gains must equal number of filters"

            # Apply each filter and gain to the audio and sum the results
            equalized_audio = sum(gain * signal.lfilter(b, a, audio) for (b, a), gain in zip(filters, gains))
            return equalized_audio

        # Load audio
        fs, audio = wavfile.read('music.wav')
        audio = audio.astype(float)

        # Equalize the audio
        gains = [1.0, 0.8, 1.2, 0.9, 1.1, 0.95] # Sample gains; adjust as necessary
        equalized_audio = equalize_audio(audio, gains)

        # Save equalized audio
        wavfile.write('equalized_music.wav', fs, equalized_audio.astype(np.int16))

        # Play the original and equalized audio for comparison
        print("Playing original audio...")
        sa.play_buffer(audio.astype(np.int16), 1, 2, fs).wait_done()
        print("Playing equalized audio...")
        sa.play_buffer(equalized_audio.astype(np.int16), 1, 2, fs).wait_done()
```