test

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```
[]: import time
     import wave
     import pyaudio
     import numpy as np
     import matplotlib.pyplot as plt
[]: FILENAME = "audio1.wav"
[]: def plot_waveform(file):
         plt.title("Waveform of the audio")
         plt.ylabel("Amplitude")
         plt.xlabel("Time")
         signal = np.frombuffer(file.readframes(-1), np.int8)
         plt.plot(signal, color="blue")
         plt.show()
[]: def play_audio(file):
         CHUNK = 1024
         # Initialize PyAudio
         audio = pyaudio.PyAudio()
         # Open stream
         stream = audio.open(format=audio.get_format_from_width(file.getsampwidth()),
                             channels=file.getnchannels(),
                             rate=file.getframerate(),
                             output=True)
         data = file.readframes(CHUNK)
         while len(data) > 0:
             stream.write(data)
             data = file.readframes(CHUNK)
         stream.stop_stream()
         stream.close()
         audio.terminate()
```

```
[]: # Function to measure latency
def measure_latency(file):
    # Record start time
    start_time = time.time()

# DSP System
    play_audio(file)

# Calculate elapsed time
    elapsed_time = time.time() - start_time

# Calculate latency in milliseconds
    latency_ms = elapsed_time * 1000
    return latency_ms
```

Latency of the RNN Digital Filter: 2172.2848415374756 milliseconds



