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In [1]:

    ★ import pyaudio

           import speech_recognition as sr
           import time
           import numpy as np
In [2]: ▶ # Audio Configuration
           CHUNK = 1024 # Audio chunks size
           FORMAT = pyaudio.paInt16 # 16-bit audio
           CHANNELS = 1 # Mono audio
           RATE = 16000 # Sampling rate (Hz)
           SILENCE_LIMIT = 1.5 # Seconds of silence to stop listening
r = sr.Recognizer()
           audio = pyaudio.PyAudio()
In [4]: ▶ # Microphone stream setup
           def get_audio_stream():
               return audio.open(
                   format=FORMAT,
                   channels=CHANNELS,
                   rate=RATE,
                   input=True,
                   frames_per_buffer=CHUNK
               )
```

In [5]:

Real-time processing

def listen_and_transcribe():

print("Listening... (Press Ctrl+C to stop)")

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stream = get_audio_stream()
                try:
                    while True:
                        print("\nSpeak now...")
                        frames = []
                        silent frames = 0
                        silence_threshold = 500 # Adjust based on your mic sensitivit
                        # Record until silence or timeout
                        for _ in range(0, int(RATE / CHUNK * 10)): # Max 10 seconds
                            data = stream.read(CHUNK, exception_on_overflow=False)
                            frames.append(data)
                            amplitude = np.frombuffer(data, dtype=np.int16).max()
                            if amplitude < silence_threshold:</pre>
                                 silent_frames += 1
                                 if silent_frames > int(SILENCE_LIMIT * RATE / CHUNK):
                                     break
                            else:
                                 silent_frames = 0
                        # Convert to audio data
                        audio_data = sr.AudioData(b''.join(frames), RATE, 2)
                        try:
                            text = r.recognize_google(audio_data)
                            print(f">>> {text}")
                        except sr.UnknownValueError:
                            print("(Could not understand audio)")
                        except sr.RequestError as e:
                            print(f"(API Error: {e})")
                except KeyboardInterrupt:
                    print("\nStopped listening")
                finally:
                    stream.stop_stream()
                    stream.close()
                    audio.terminate()
In [6]:
            # Start listening
            listen_and_transcribe()
            Listening... (Press Ctrl+C to stop)
            Speak now...
            >>> hello what are you doing
            Speak now...
            (Could not understand audio)
            Speak now...
            Stopped listening
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