NOISE POLLUTION PYTHON SCRIPT

Program No:1

```
import pyaudio
import numpy as np
import sounddevice as sd
import soundfile as sf
# Parameters for audio recording
sample rate = 44100 # Sample rate (samples per second)
duration = 10 # Duration of recording (in seconds)
# Parameters for noise level threshold
threshold_db = 40 # Set your desired threshold in decibels
def record_audio(duration, sample_rate):
  audio_data = np.array([], dtype=np.int16)
  print("Recording audio...")
  # Initialize audio recording
  with sd.InputStream(callback=callback):
```

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sd.sleep(duration * 1000)
  return audio_data
def callback(indata, frames, time, status):
  if status:
    print(status, file=sys.stderr)
  if any(indata):
    # Process audio data here if needed
    audio_data = indata
    # You can add your noise pollution analysis code here
if __name__ == "__main__":
  audio data = record audio(duration, sample rate)
  # Analyze the audio data and calculate the noise level
  noise_level = np.max(audio_data) # You can replace this with your own noise
analysis algorithm
  print(f"Noise level: {noise_level} dB")
 if noise_level > threshold_db:
    print("Noise pollution detected!")
  else:
    print("No significant noise pollution detected.")
```

Program No:2

```
import pyaudio
import numpy as np
# Constants for audio settings
FORMAT = pyaudio.paInt16
CHANNELS = 1
RATE = 44100 # Sample rate (samples per second)
CHUNK = 1024 # Size of each audio chunk
THRESHOLD = 2000 # Adjust this threshold as needed
# Initialize the audio stream
p = pyaudio.PyAudio()
stream = p.open(format=FORMAT,
        channels=CHANNELS,
        rate=RATE,
        input=True,
        frames_per_buffer=CHUNK)
print("Listening...")
```

```
try:
  while True:
    data = stream.read(CHUNK)
    audio_data = np.frombuffer(data, dtype=np.int16)
    rms = np.sqrt(np.mean(audio_data**2))
    if rms > THRESHOLD:
      print(f"Noise level: {rms:.2f} dB")
except KeyboardInterrupt:
  print("Recording stopped.")
finally:
  stream.stop_stream()
  stream.close()
  p.terminate()
import soundmeter
import time
```

Program No:3

```
# Initialize the sound meter
meter = soundmeter.Meter()
# Create a log file to store noise level data
log_file = 'noise_log.txt'
try:
  while True:
    # Measure the noise level
    noise_level = meter.get_level()
    # Get the current timestamp
    timestamp = time.strftime("%Y-%m-%d %H:%M:%S")
    # Print the noise level and save it to the log file
    print(f"{timestamp} - Noise Level: {noise_level} dB")
    with open(log_file, 'a') as f:
      f.write(f"{timestamp} - Noise Level: {noise_level} dB\n")
```

```
# Sleep for a specified interval (e.g., 1 minute)
    time.sleep(60)
except KeyboardInterrupt:
  pass
print("Monitoring stopped.")
Program No:4
import sounddevice as sd
import numpy as np
import math
def calculate_noise_level(audio_data, sample_rate):
  # Calculate the FFT of the audio data
  fft_data = np.fft.fft(audio_data)
  num_samples = len(audio_data)
  # Calculate the frequency values for each FFT bin
  frequencies = np.fft.fftfreq(num_samples, 1.0 / sample_rate)
```

```
# Find the peak frequency and its corresponding amplitude
  peak_freq_index = np.argmax(np.abs(fft_data))
  peak freg = abs(frequencies[peak freg index])
  peak_amplitude = abs(fft_data[peak_freq_index])
  # Calculate the noise level in decibels (dB)
  noise_level_dB = 20 * math.log10(peak_amplitude)
  return noise level dB
def main():
  duration = 10 # Duration of the recording in seconds
  sample rate = 44100 # Sampling rate in Hz
  print("Recording... (Press Ctrl+C to stop)")
  try:
    audio_data = sd.rec(int(duration * sample_rate), samplerate=sample_rate,
channels=1, dtype='float64')
    sd.wait()
    noise level = calculate noise level(audio data, sample rate)
```

```
print(f"Noise Level: {noise_level:.2f} dB")
except KeyboardInterrupt:
    print("\nRecording stopped.")

if __name__ == "__main__":
    main()
```