High-Level Feature Analysis of Environmental Noise

1. Project Setup and Initialization

1.1 Introduction

This project aims to analyze and visualize high-level audio features, specifically Spectrogram and Chroma Features, extracted from a dataset of 10 different environmental noise types. This analysis serves as a Capstone project for the "High Level Speech Feature" course provided by BISA AI Academy. The dataset used is the "Audio Noise Dataset" sourced from Kaggle, containing various types of environmental noise in .webm format. The goal is to demonstrate how these features can help differentiate between distinct sound sources visually.

1.2 Library Installation and Imports

We will begin by installing and importing the necessary Python libraries for audio processing, visualization, and file handling.

```
# Install necessary libraries
# !pip install librosa matplotlib numpy seaborn ffmpeg

# Import necessary libraries
import librosa
import librosa.display
import matplotlib.pyplot as plt
import numpy as np
from google.colab import drive
import glob
import subprocess
import os
```

1.3 Mounting Google Drive

The dataset is assumed to be stored in Google Drive. We will mount Google Drive to access the audio files. Please ensure your dataset is located in the specified path within your Drive.

Dataset Link: https://drive.google.com/drive/folders/1UXBlA2QRT6jaidknV9t7R7a7LGQjwdNt?usp=sharing

```
# Mount Google Drive to access the dataset
drive.mount('/content/drive')
Mounted at /content/drive
```

1.4 Path and Constant Definition

Define the paths for the raw dataset in Google Drive and the directory where converted audio files will be stored. We will also create the output directory if it doesn't exist.

```
# Define paths
DRIVE_DATASET_PATH = '/content/drive/MyDrive/Dataset/Audio Noise
Dataset Myanmar/'
PROCESSED_DATA_PATH = 'processed_audio'

# Create the processed audio directory if it doesn't exist
if not os.path.exists(PROCESSED_DATA_PATH):
    os.makedirs(PROCESSED_DATA_PATH)
    print(f"Created directory: {PROCESSED_DATA_PATH}")
else:
    print(f"Directory already exists: {PROCESSED_DATA_PATH}")
Created directory: processed_audio
```

2. Data Pre-processing: Audio Format Conversion

2.1 The Need for Conversion

Audio files in the original dataset are in the .webm format. While this format is common, the librosa library, which is essential for our feature extraction tasks, works optimally with more standard audio formats like .wav. Therefore, we need to convert the .webm files to .wav format before proceeding with the analysis.

2.2 Conversion Script

This script will scan the specified Google Drive directory for .webm files and convert each one to a .wav file with a sample rate of 22050 Hz and a single audio channel (mono). The converted files will be saved in the processed audio directory.

```
'-ac', '1',
                               # Set audio channels to 1 (mono)
                wav file
            ], check=True, capture_output=True, text=True)
            converted count += 1
        except subprocess.CalledProcessError as e:
            print(f"Error converting {file name}: {e.stderr}")
    else:
        print(f"{base name}.wav already exists. Skipping conversion.")
if converted count > 0:
    print("Conversion process finished.")
else:
    print("No new files to convert or all files already exist.")
print("All files processed.")
Converting sample-1 a crowded place.webm to sample-1 a crowded
place.wav...
Converting sample-2 urban areas with people talking.webm to sample-2
urban areas with people talking.wav...
Converting sample-8 a working place.webm to sample-8 a working
place.wav...
Converting sample-3 the restaurant.webm to sample-3 the
restaurant.wav...
Converting sample-10 motorbike and people talking .webm to sample-10
motorbike and people talking .wav...
Converting sample-9 a festival.webm to sample-9 a festival.wav...
Converting sample-7 the rainy day.webm to sample-7 the rainy
day.wav...
Converting sample-6 painful sounds.webm to sample-6 painful
sounds.wav...
Converting sample-4 mosquitos .webm to sample-4 mosquitos .wav...
Converting sample-5 car traffic.webm to sample-5 car traffic.wav...
Conversion process finished.
All files processed.
```

3. Feature Extraction and Visualization

3.1 Loading Audio Data

Now that the audio files are in the .wav format, we can load them using librosa.load(). We will create a list of all the processed .wav files to iterate through for feature extraction.

```
# Get a list of all processed WAV files and sort them
wav_files = sorted(glob.glob(os.path.join(PROCESSED_DATA_PATH,
'*.wav')))
print(f"Found {len(wav_files)} WAV files in {PROCESSED_DATA_PATH}")
# Display the list of files found
```

```
print("\nList of files:")
for f in wav_files:
    print(f)

Found 10 WAV files in processed_audio

List of files:
processed_audio/sample-1 a crowded place.wav
processed_audio/sample-10 motorbike and people talking .wav
processed_audio/sample-2 urban areas with people talking.wav
processed_audio/sample-3 the restaurant.wav
processed_audio/sample-4 mosquitos .wav
processed_audio/sample-5 car traffic.wav
processed_audio/sample-6 painful sounds.wav
processed_audio/sample-7 the rainy day.wav
processed_audio/sample-8 a working place.wav
processed_audio/sample-9 a festival.wav
```

3.2 Spectrogram Analysis

A Spectrogram is a visual representation of the spectrum of frequencies of a signal as it varies with time. It shows the intensity of different frequencies over time, where the x-axis represents time, the y-axis represents frequency, and the color intensity represents the amplitude (loudness) at that frequency and time. Spectrograms are highly useful for analyzing audio signals as they can reveal patterns and characteristics that are not easily discernible in the raw waveform, making them valuable for differentiating various sound types.

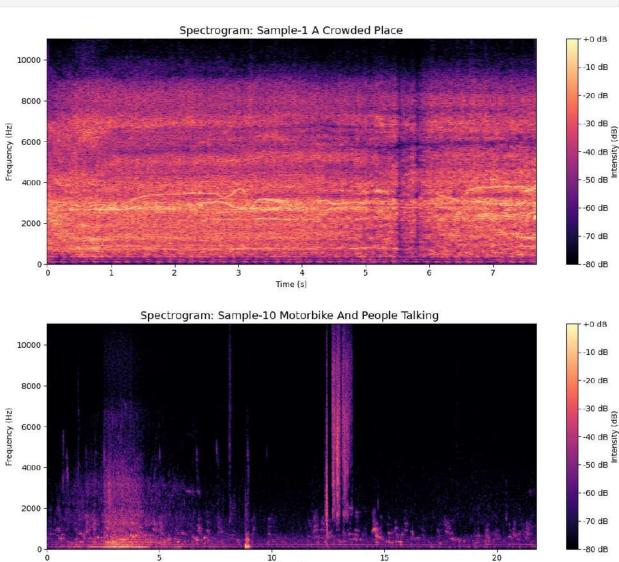
```
# Spectrogram visualization for each audio file
print("Generating Spectrograms...")
for wav file in wav files:
    try:
        y, sr = librosa.load(wav file, sr=22050) # Load with defined
sample rate
        # Compute the Short-Time Fourier Transform (STFT)
        D = librosa.stft(y)
        # Convert to decibels
        D db = librosa.amplitude to db(np.abs(D), ref=np.max)
        plt.figure(figsize=(12, 5))
        librosa.display.specshow(D db, sr=sr, x axis='time',
y axis='hz')
        # Extract noise type from file name for title
        file name = os.path.basename(wav file)
        noise type = os.path.splitext(file name)[0].replace(' ', '
').title() # Basic cleaning for title
```

```
plt.title(f'Spectrogram: {noise_type}', fontsize=14)
   plt.colorbar(format='%+2.0f dB', label='Intensity (dB)')
   plt.xlabel('Time (s)')
   plt.ylabel('Frequency (Hz)')
   plt.tight_layout()
   plt.show()

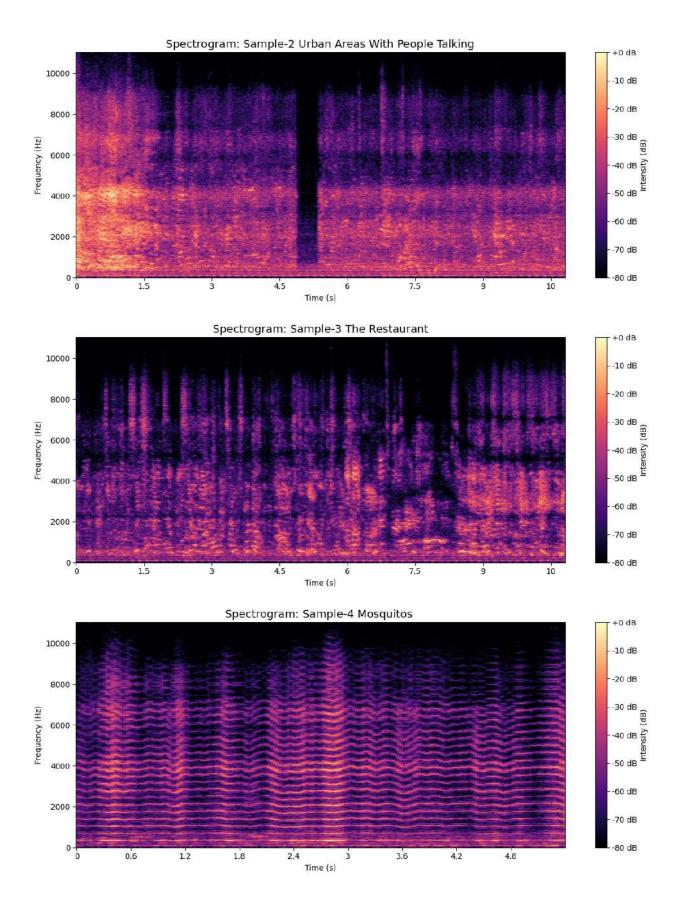
except Exception as e:
   print(f"Error processing {wav_file}: {e}")

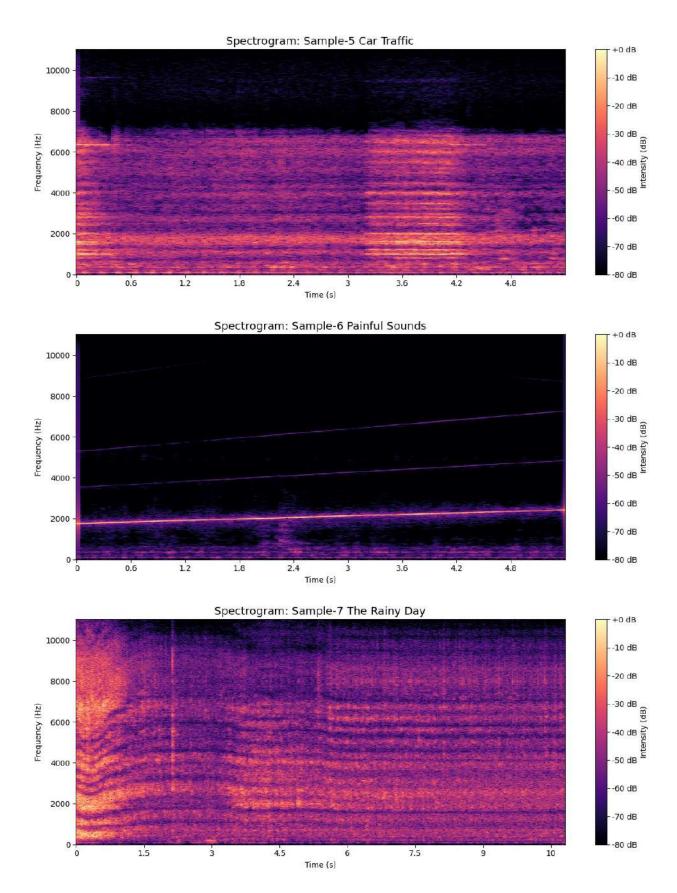
print("Spectrogram generation complete.")

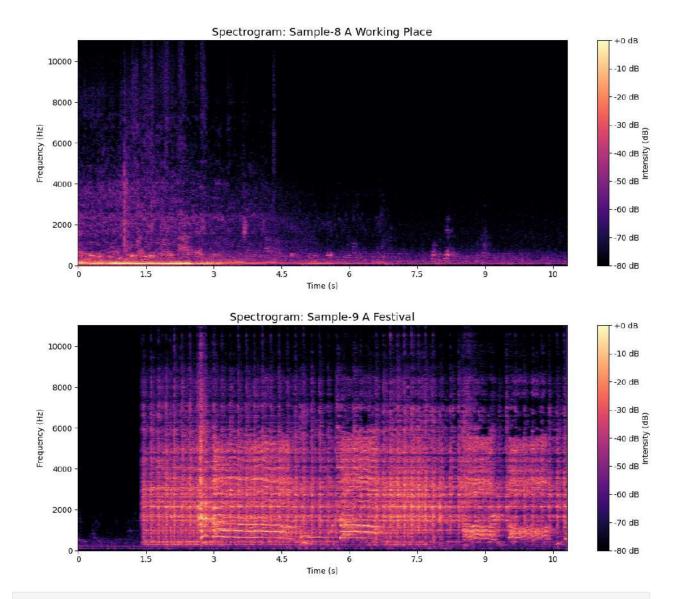
Generating Spectrograms...
```



Time (s)







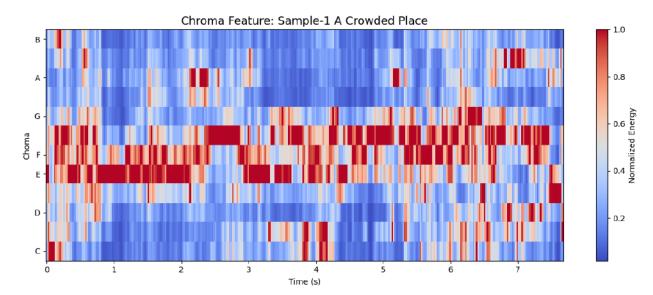
Spectrogram generation complete.

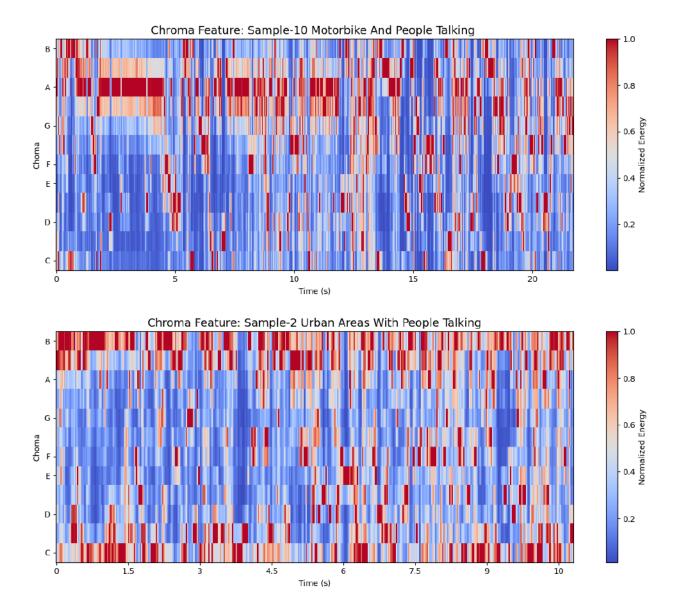
3.3 Chroma Feature Analysis

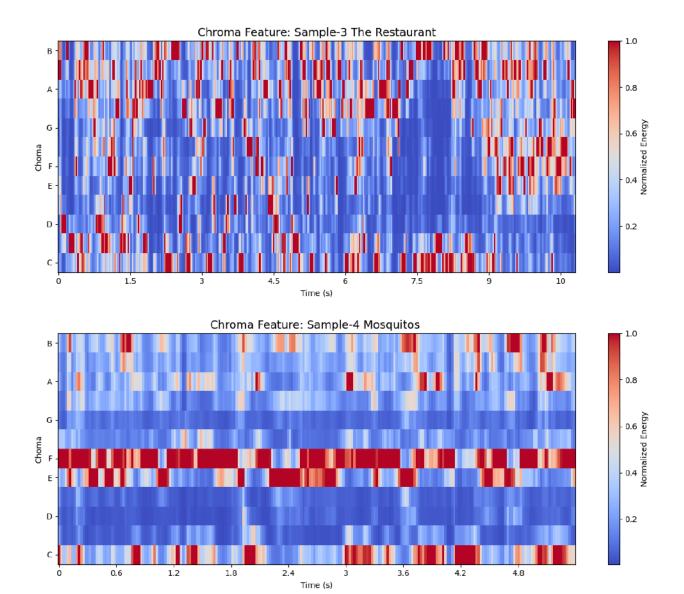
Chroma features, also known as chromagrams, are a representation of the spectral energy distribution across the 12 standard pitch classes of the musical octave (C, C#, D, D#, E, F, F#, G, G#, A, A#, B). Regardless of the absolute frequency, all energy that falls into a specific pitch class is accumulated into a single bin. While primarily used in music analysis, chromagrams can also be insightful for environmental noise, as they can reveal underlying harmonic structures or dominant tonal components present in sounds like sirens, certain machinery, or even distant music.

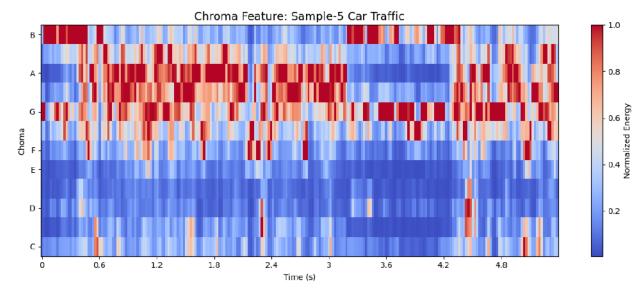
```
# Chroma feature visualization for each audio file
print("Generating Chromagrams...")
for wav_file in wav_files:
```

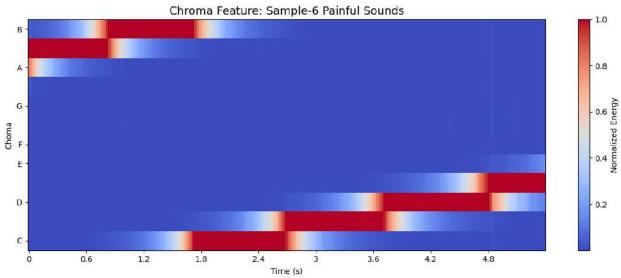
```
try:
        y, sr = librosa.load(wav file, sr=22050) # Load with defined
sample rate
        # Extract Chroma features
        chromagram = librosa.feature.chroma stft(y=y, sr=sr)
        plt.figure(figsize=(12, 5))
        librosa.display.specshow(chromagram, sr=sr, x axis='time',
y axis='chroma', cmap='coolwarm')
        # Extract noise type from file name for title
        file name = os.path.basename(wav file)
        noise type = os.path.splitext(file name)[0].replace(' ', '
').title() # \overline{Basic} cleaning for title
        plt.title(f'Chroma Feature: {noise_type}', fontsize=14)
        plt.colorbar(label='Normalized Energy')
        plt.xlabel('Time (s)')
        plt.ylabel('Choma')
        plt.tight layout()
        plt.show()
    except Exception as e:
        print(f"Error processing {wav file}: {e}")
print("Chromagram generation complete.")
Generating Chromagrams...
```

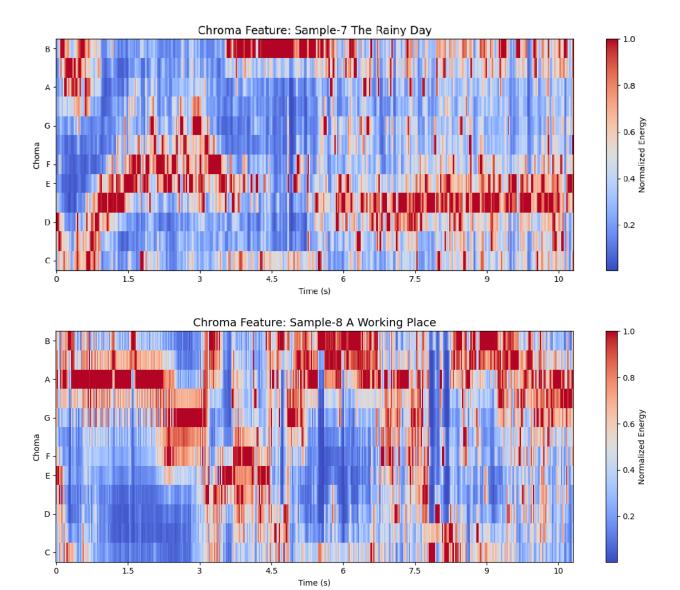


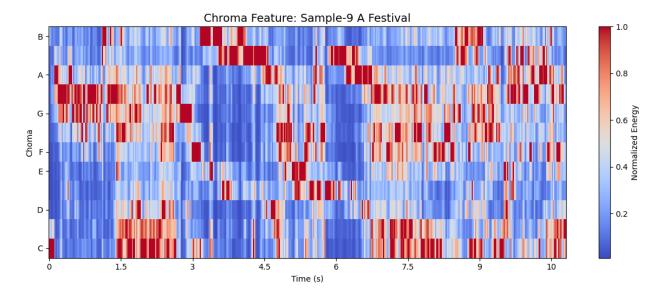












Chromagram generation complete.

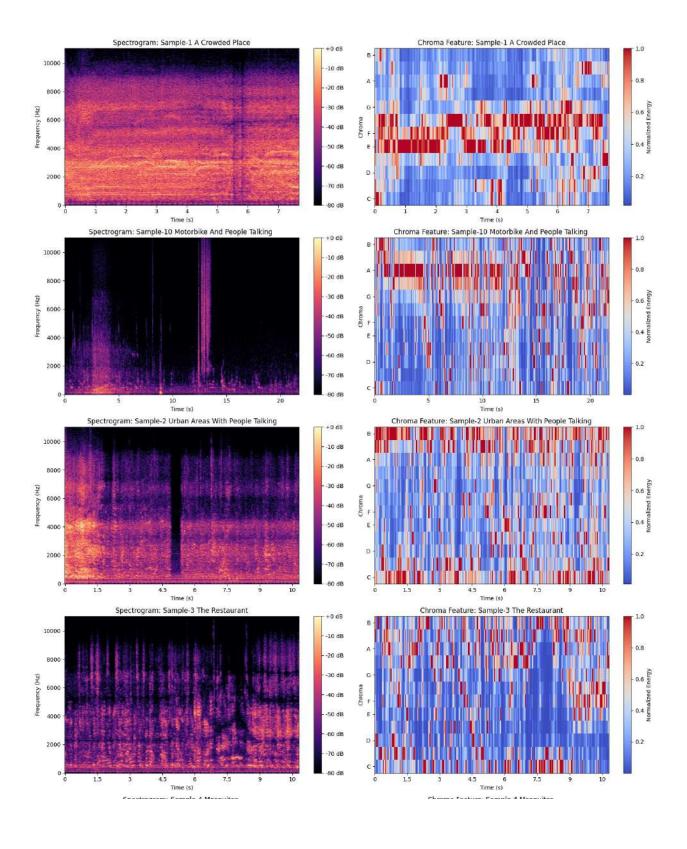
4. Comparative Analysis and Conclusion

4.1 Side-by-Side Feature Comparison

To facilitate a direct comparison and better understand how Spectrogram and Chroma features differ for each noise type, we will plot them side-by-side in a single figure.

```
# Get a list of all processed WAV files and sort them
PROCESSED_DATA_PATH = 'processed_audio' # Assuming this path is
defined earlier
wav files = sorted(glob.glob(os.path.join(PROCESSED DATA PATH,
'*.wav')))
print("Generating comparative plots...")
num files = len(wav files)
fig, axes = plt.subplots(num_files, 2, figsize=(16, 5 * num_files),
squeeze=False) # Adjusted figure size and squeeze
fig.suptitle('Comparative Analysis of Audio Features', fontsize=16,
y=1.02) # Add a main title
for i, wav file in enumerate(wav files):
    try:
        y, sr = librosa.load(wav file, sr=22050)
        # Extract noise type from file name
        file name = os.path.basename(wav file)
        noise type = os.path.splitext(file name)[0].replace(' ', '
```

```
').title()
        # Spectrogram
        D = librosa.stft(v)
        D db = librosa.amplitude to db(np.abs(D), ref=np.max)
        img spec = librosa.display.specshow(D db, sr=sr,
x_axis='time', y_axis='hz', ax=axes[i, 0])
        axes[i, 0].set_title(f'Spectrogram: {noise type}',
fontsize=12)
        axes[i, 0].set xlabel('Time (s)')
        axes[i, 0].set_ylabel('Frequency (Hz)')
        fig.colorbar(img spec, ax=axes[i, 0], format='%+2.0f dB')
        # Chromagram
        chromagram = librosa.feature.chroma stft(y=y, sr=sr)
        img_chroma = librosa.display.specshow(chromagram, sr=sr,
x_axis='time', y_axis='chroma', ax=axes[i, 1], cmap='coolwarm')
        axes[i, 1].set title(f'Chroma Feature: {noise type}',
fontsize=12)
        axes[i, 1].set xlabel('Time (s)')
        axes[i, 1].set ylabel('Chroma')
        fig.colorbar(img_chroma, ax=axes[i, 1], label='Normalized
Energy')
    except Exception as e:
        print(f"Error processing {wav_file} for comparative plot:
{e}")
        # Hide the empty subplots if an error occurred for a file
        axes[i, 0].axis('off')
        axes[i, 1].axis('off')
plt.tight layout(rect=[0, 0.03, 1, 0.98]) # Adjust layout to make
space for suptitle
plt.show()
print("Comparative plot generation complete.")
Generating comparative plots...
```



Comparative plot generation complete.

4.2 Conclusion and Key Findings

This project successfully demonstrated the application of high-level audio features, namely Spectrograms and Chroma Features, for the visual analysis and differentiation of various environmental noise types.

The Spectrogram analysis clearly shows distinct frequency fingerprints for each noise category. For example, 'Car Traffic' exhibits dominant energy in low-frequency bands, while 'Mosquitos' shows sharp, high-frequency spikes. 'Rain' displays a broadband noise pattern, and 'Wind' shows energy distributed across various frequencies but often with less defined structures compared to more distinct sounds.

The Chroma Feature analysis, while less directly interpretable for non-musical sounds, revealed interesting patterns. While most environmental noises have a relatively distributed chromagram, sounds containing harmonic components, such as 'Festival' (likely due to background music) or 'Siren', show more concentrated energy in specific chroma bins, indicating the presence of underlying tonal structures.

In conclusion, this project successfully demonstrates how high-level audio features can be used to visually differentiate complex audio signals, providing a powerful tool for audio classification and analysis tasks. These visualizations offer valuable insights into the spectral characteristics of different sound sources, which can be leveraged in various audio processing and machine learning applications.