

## IMPORTS

```
In [ ]: import os
import math
import librosa
import librosa.display
import numpy as np
import matplotlib.pyplot as plt
import numpy as np
from sklearn.decomposition import PCA
from sklearn.preprocessing import StandardScaler
import joblib
from scipy.signal import butter, lfilter
import soundfile as sf
```

## VARIABLES GLOBALES

```
In [ ]: fruit_types      = ['pera', 'banana', 'manzana', 'naranja']
audios      = {fruit: [] for fruit in fruit_types}
training_path = '../.../dataset/audios/training'
original_path = os.path.join(training_path, 'original')
processed_path = os.path.join(training_path, 'processed')
model_file    = 'model.pkl'
model         = dict.fromkeys(['pca', 'features', 'scaler'])
```

## DICCIONARIO DE AUDIOS ORIGINALES

```
In [ ]: original = {fruit: [] for fruit in fruit_types}
for dirname, _, filenames in os.walk(original_path):
    subdir = os.path.basename(dirname)
    if subdir in fruit_types:
        original[subdir].extend([os.path.join(dirname, filename) for filename in filenames])
```

## DICCIONARIO DE AUDIOS PROCESADOS

```
In [ ]: processed = {fruit: [] for fruit in fruit_types}
for dirname, _, filenames in os.walk(processed_path):
    subdir = os.path.basename(dirname)
    if subdir in fruit_types:
        processed[subdir].extend([os.path.join(dirname, filename) for filename in filenames])
```

## PARAMETROS DEL AUDIO

```
In [ ]: FRAME_SIZE = 512# In the documentation says it's convenient for speech.C
HOP_SIZE    = int(FRAME_SIZE/2)
```

## FUNCIONES GENERALES DE AUDIO

```
In [ ]: def load_audio(audiofile):
        test_audio, sr = librosa.load(audiofile, sr = None)
        duration = librosa.get_duration(filename=audiofile, sr=sr)
        return test_audio, sr, duration
```

## FILTERS

```
In [ ]: def band_pass_filter(signal, sr, low_cutoff, high_cutoff):
        b, a = butter(N=3, Wn = [low_cutoff, high_cutoff], btype='band', fs=sr)
        return lfilter(b, a, signal)
```

## PRROCESSING OF THE AUDIO FILES FUNCTIONS

```
In [ ]: def spectral_flux(signal):

        # Calcular el espectrograma de magnitudes
        spectrogram = np.abs(librosa.stft(signal, n_fft = FRAME_SIZE, hop_length=HOP_SIZE))

        # Calcular el flujo espectral
        spectral_flux_values = np.sum(np.diff(spectrogram, axis=1)**2, axis=0)

        return spectral_flux_values
```

```
In [ ]: def process(audio_in, audio_out, rms_umbral = 0.043, flux_umbral = 0.096):
        signal, sr, _ = load_audio(audio_in)

        rms = librosa.feature.rms(y = signal, frame_length = FRAME_SIZE, hop_length=HOP_SIZE)
        rms /= np.max(np.abs(rms))
        trms = librosa.times_like(rms, sr = sr, hop_length = HOP_SIZE, n_fft = FRAME_SIZE)
        trms /= trms[-1]

        flux = spectral_flux(signal)
        flux /= np.max(np.abs(flux))
        fluxframes = range(len(flux))
        tflux = librosa.frames_to_time(fluxframes, hop_length=HOP_SIZE, n_fft = FRAME_SIZE)
        tflux /= tflux[-1]

        left_index = np.argmax(np.abs(flux) > flux_umbral)
        righth_index = len(flux) - 1 - np.argmax(np.abs(np.flip(flux))) > flux_umbral

        tsignal = librosa.times_like(signal, sr = sr, hop_length=HOP_SIZE, n_fft = FRAME_SIZE)
        tsignal /= tsignal[-1]

        flag = False
        pad_left = 0
        pad_righth = 0
        flag_left = False
        flag_righth = False

        while not flag:
            if rms[0, left_index] > rms_umbral:
                if left_index > pad_left + 15:
```

```

        rms_left = left_index - np.argmax(np.flip(np.abs(rms[0, :1
    if rms_left <= 0:
        rms_left = left_index
    flag_left = True
    else:
        pad_left += 15
        left_index = pad_left + np.argmax(np.abs(flux[pad_left:]))
    else:
        rms_left = left_index
        flag_left = True

    if rms[0, righth_index] > rms_umbral:
        if righth_index < (len(flux) - 1 - pad_righth-15):
            rms_righth = righth_index + np.argmax(np.abs(rms[0, righth_in
            if rms_righth >= len(flux):
                rms_righth = righth_index
            flag_righth = True
        else:
            pad_righth += 15
            righth_index = len(flux[:-pad_righth]) - 1 - np.argmax(np.fl
    else:
        rms_righth = righth_index
        flag_righth = True

    flag = flag_left and flag_righth

left_index = min(left_index, rms_left)
righth_index = max(righth_index, rms_righth)

mask = tsignal >= tflux[left_index]
ttrimmed = tsignal[mask]
trimmed = signal[mask]
mask = ttrimmed <= tflux[righth_index]
ttrimmed = ttrimmed[mask]
trimmed = trimmed[mask]
sf.write(audio_out, trimmed, sr)

```

```

In [ ]: def process_audios(original:dict, processed:dict):
    already_processed = []
    for group in processed.values():
        already_processed.extend([os.path.basename(audio) for audio in gro

    for fruit, audios in original.items():
        for audio in audios:
            file = os.path.basename(audio)
            if file in already_processed:
                pass
            else:
                audio_out = os.path.join(processed, f"{fruit}/{file}")
                process(audio, audio_out)
                processed[fruit].append(audio_out)

```

## PLOTTING

```
In [ ]: #3d
def plot_features3d(features):
    fig = plt.figure()
    ax = fig.add_subplot(111, projection='3d')
    colors = dict(zip(fruit_types, ['green', 'yellow', 'red', 'orange']))

    for fruit, points in features.items():
        ax.scatter(points[:, 0], points[:, 1], points[:, 2], c=colors[fruit])

    ax.set_xlabel('Eje X')
    ax.set_ylabel('Eje Y')
    ax.set_zlabel('Eje Z')
    plt.show()
```

## AUDIO PROCESSING

```
In [ ]: process_audios(original, processed)
```

## FEATURES EXTRACTION

*Features extraction functions*

```
In [ ]: def calculate_split_frequency_bin(split_frequency, sample_rate, num_frequency_bins):
    """Infer the frequency bin associated to a given split frequency."""

    frequency_range = sample_rate / 2
    frequency_delta_per_bin = frequency_range / num_frequency_bins
    split_frequency_bin = math.floor(split_frequency / frequency_delta_per_bin)
    return int(split_frequency_bin)
```

```
In [ ]: def band_energy_ratio(spectrogram, split_frequency, sample_rate):
    """Calculate band energy ratio with a given split frequency."""

    split_frequency_bin = calculate_split_frequency_bin(split_frequency, sample_rate, num_frequency_bins)
    band_energy_ratio = []

    # calculate power spectrogram
    power_spectrogram = np.abs(spectrogram) ** 2
    power_spectrogram = power_spectrogram.T

    # calculate BER value for each frame
    for frame in power_spectrogram:
        sum_power_low_frequencies = frame[:split_frequency_bin].sum()
        sum_power_high_frequencies = frame[split_frequency_bin:].sum()
        band_energy_ratio_current_frame = sum_power_low_frequencies / (sum_power_low_frequencies + sum_power_high_frequencies)
        band_energy_ratio.append(band_energy_ratio_current_frame)

    return np.array(band_energy_ratio)
```

```
In [ ]: def rms(signal, frames, hop):
    return librosa.feature.rms(y=signal, frame_length = frames, hop_length = hop)
```

*function to get the features*

```
In [ ]: def get_features(signal, sr):
        feature = np.empty((1, 0))

        # BER
        spec = librosa.stft(signal, n_fft = FRAME_SIZE, hop_length = HOP_SIZE)
        # max
        split_frequency = 600
        BER = band_energy_ratio(spec, split_frequency, sr)
        feat = np.max(np.abs(BER))
        feature = np.append(feature, feat)
        # min
        # 1
        split_frequency = 1900
        BER = band_energy_ratio(spec, split_frequency, sr)
        feat = np.min(np.abs(BER))
        feature = np.append(feature, feat)
        # 2
        split_frequency = 5000
        BER = band_energy_ratio(spec, split_frequency, sr)
        feat = np.min(np.abs(BER))
        feature = np.append(feature, feat)
        # 3
        split_frequency = 9000
        BER = band_energy_ratio(spec, split_frequency, sr)
        feat = np.min(np.abs(BER))
        feature = np.append(feature, feat)
        # std
        # 1
        split_frequency = 8000
        BER = band_energy_ratio(spec, split_frequency, sr)
        BER /= np.max(np.abs(BER))
        feat = np.std(BER)/np.mean(np.abs(BER))
        feature = np.append(feature, feat)
        # 2
        split_frequency = 1000
        BER = band_energy_ratio(spec, split_frequency, sr)
        BER /= np.max(np.abs(BER))
        feat = np.std(BER)/np.mean(np.abs(BER))
        feature = np.append(feature, feat)

        #ZCR
        cutoff = 5000
        cuton = 1000
        filtered = band_pass_filter(signal, sr, cuton, cutoff)
        zcr = librosa.feature.zero_crossing_rate(filtered, frame_length=FRAME_
        zcr /= np.max(np.abs(zcr))
        # mean
        feat = np.mean(zcr)
        feature = np.append(feature, feat)
        # maximum
        cutoff = 10000
```

```

cuton = 10
filtered = band_pass_filter(signal, sr, cuton, cutoff)
zcr = librosa.feature.zero_crossing_rate(filtered, frame_length=FRAME_
feat = np.max(np.abs(zcr))
feature = np.append(feature, feat)
# std
cutoff = 10000
cuton = 20
filtered = band_pass_filter(signal, sr, cuton, cutoff)
zcr = librosa.feature.zero_crossing_rate(filtered, frame_length=FRAME_
feat = np.std(zcr)/np.mean(np.abs(zcr))
feature = np.append(feature, feat)
# mean local
cutoff = 5000
cuton = 1000
filtered = band_pass_filter(signal, sr, cuton, cutoff)
zcr = librosa.feature.zero_crossing_rate(filtered, frame_length=FRAME_
zcr /= np.max(np.abs(zcr))
feat = np.mean(zcr[((len(zcr)*3)//14 - 5) : ((len(zcr)*3)//14 + 5)])
feature = np.append(feature, feat)
# local max
cutoff = 10000
cuton = 10
filtered = band_pass_filter(signal, sr, cuton, cutoff)
zcr = librosa.feature.zero_crossing_rate(filtered, frame_length=FRAME_
feat = np.max(zcr[((len(zcr)*3)//4 - 10) : ((len(zcr)*3)//4 + 10)])
feature = np.append(feature, feat)

# Roll off
cuton = 100
cutoff = 8500
filtered = band_pass_filter(signal, sr, cuton, cutoff)
roll_off = librosa.feature.spectral_rolloff(y=filtered, sr=sr, n_fft=F
roll_off /= np.max(np.abs(roll_off))
# mean
feat = np.mean(np.abs(roll_off))
feature = np.append(feature, feat)
# max
cuton = 100
cutoff = 8500
filtered = band_pass_filter(signal, sr, cuton, cutoff)
roll_off = librosa.feature.spectral_rolloff(y=filtered, sr=sr, n_fft=F
feat = np.max(np.abs(roll_off))
feature = np.append(feature, feat)
# std
cutoff = 8500
cuton = 50
filtered = band_pass_filter(signal, sr, cuton, cutoff)
roll_off = librosa.feature.spectral_rolloff(y=filtered, sr=sr, n_fft=F
roll_off /= np.max(np.abs(roll_off))
feat = np.std(np.abs(roll_off))/np.mean(np.abs(roll_off))
feature = np.append(feature, feat)

#MFCCS

```

```

n_mfcc = 4
# 1
cuton = 500
cutoff = 5000
filtered = band_pass_filter(signal, sr, cuton, cutoff)
mfccs = librosa.feature.mfcc(y = signal, sr=sr, n_mfcc = n_mfcc, n_fft=
feat = np.max(mfccs, axis = 1)
feature = np.append(feature, feat[3])

# 2
cuton = 10
cutoff = 8000
filtered = band_pass_filter(signal, sr, cuton, cutoff)
mfccs = librosa.feature.mfcc(y = filtered, sr=sr, n_mfcc = n_mfcc, n_f
mfccs /= np.max(np.abs(mfccs), axis = 1, keepdims=True)
feat = np.std(np.abs(mfccs), axis = 1)/np.mean(np.abs(mfccs), axis = 1)
feature = np.append(feature, feat[3])

# 3
cuton = 10
cutoff = 8000
filtered = band_pass_filter(signal, sr, cuton, cutoff)
mfccs = librosa.feature.mfcc(y = filtered, sr=sr, n_mfcc = n_mfcc, n_f
mfccs /= np.max(np.abs(mfccs), axis = 1, keepdims=True)
mfccs = mfccs[:, ((mfccs.shape[1]*4) // 5 - 10):((mfccs.shape[1]*4) //
feat = np.std(np.abs(mfccs), axis=1) / np.mean(np.abs(mfccs), axis=1)
feature = np.append(feature, feat[1])

#envelope
env = rms(signal, FRAME_SIZE, HOP_SIZE)
env = env.reshape(-1,)
selected = np.linspace(0, len(env) - 1, 30, dtype=int)
env = env[selected]
feat = env[11]
feature = np.append(feature, feat)
feat = env[12]
feature = np.append(feature, feat)

return feature

```

## EXTRACTION OF FEATURES FROM PROCESSED AUDIOS

```

In [ ]: def extract_features(processed:dict):
    features = dict.fromkeys(fruit_types)
    for fruit, audios in processed.items():
        features[fruit] = None

        for audio in audios:
            # Load the audio signal
            signal, sr, _ = load_audio(audio)
            feature = get_features(signal, sr)

            if features[fruit] is not None:

```

```

        features[fruit] = np.vstack([features[fruit], feature])
    else:
        features[fruit] = feature
    return features

```

## TRAINING AUDIOS FEATURES EXTRACTION

```
In [ ]: features = extract_features(processed)
```

C:\Users\Juan\AppData\Local\Temp\ipykernel\_3228\1587896153.py:3: FutureWarning: get\_duration() keyword argument 'filename' has been renamed to 'path' in version 0.10.0.

This alias will be removed in version 1.0.

```

    duration = librosa.get_duration(filename=audiofile, sr=sr)
c:\Users\Juan\AppData\Local\Programs\Python\Python311\Lib\site-packages\librosa\feature\spectral.py:2143: UserWarning: Empty filters detected in mel frequency basis. Some channels will produce empty responses. Try increasing your sampling rate (and fmax) or reducing n_mels.
    mel_basis = filters.mel(sr=sr, n_fft=n_fft, **kwargs)

```

## PCA

```
In [ ]: #PCA and dump
whole = np.concatenate(list(features.values()), axis=0)

#Aplicar PCA para obtener dos componentes principales
pca = PCA(n_components = 3)
scaler = StandardScaler()
whole_scaled = scaler.fit_transform(whole)
reduced_features = pca.fit_transform(whole_scaled)

```

## REDUCED MODEL

```
In [ ]: #Paso 3: Crear un diccionario con las matrices reducidas
reduced = {}
start_idx = 0

for fruit, matrix in features.items():
    num_rows = matrix.shape[0]
    reduced[fruit] = reduced_features[start_idx:start_idx + num_rows, :]
    start_idx += num_rows

```

## DUMPING TO FILE

```
In [ ]: model['pca'] = pca
model['features'] = reduced
model['scaler'] = scaler
joblib.dump(model, model_file)

```

```
Out[ ]: ['model.pkl']
```

## PLOTTING



```
In [ ]: plot_features3d(reduced)
```

