

AASD 4004 Machine Learning - II

Applied Al Solutions Developer Program



Module 16 Audio Use Cases

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Agenda

Audio Classification
Unsupervised Audio Segmentation
Speaker Diarization
Speech to Text
Text to Speech





Classify audio clips into different musical genres

WAV files are organized in two folders representing two genres

Audio features to be extracted, and a classifier model to be built on these features



<u>Steps</u>

Load the WAV files from folders into Numpy arrays

Extract segment-level feature statistics from them

Choose any classifier algorithm (SVM)

Train a SVM model

Apply the model on the data

Plot the predictions (decision boundary to split two genres)



Load the WAV files from folders into Numpy arrays

```
dirs = ["data/music/classical", "data/music/metal"]
class_names = [os.path.basename(d) for d in dirs]
```

Extract segment-level feature statistics from them

```
features = []
for d in dirs: # get feature matrix for each directory (class)
    f, files, fn = aF.directory_feature_extraction(d, m_win, m_step, s_win, s_step)
    features.append(f)
```



Train a SVM model

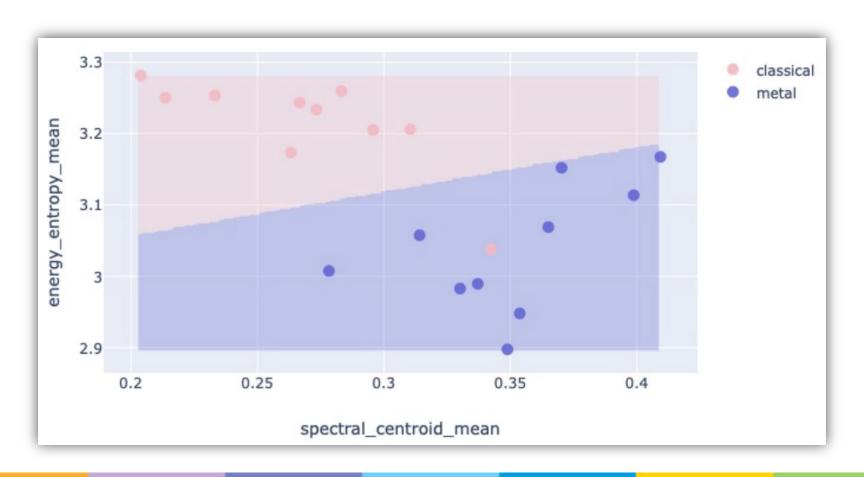
```
cl = SVC(kernel='rbf', C=20)
cl.fit(f, y)
```

Apply the model on the data

```
x_ = np.arange(f[:, 0].min(), f[:, 0].max(), 0.002)
y_ = np.arange(f[:, 1].min(), f[:, 1].max(), 0.002)
xx, yy = np.meshgrid(x_, y_)
Z = cl.predict(np.c_[xx.ravel(), yy.ravel()]).reshape(xx.shape) / 2
```



Plot the predictions (decision boundary to split two genres)





PyAudioAnalysis Wrapper - Feature Extraction + Classifier

```
from pyAudioAnalysis.audioTrainTest import extract_features_and_train
mt, st = 1.0, 0.05
dirs = ["data/music/classical", "data/music/metal"]
extract_features_and_train(dirs, mt, mt, st, st, "svm_rbf", "svm_classical_metal")
```



Best Classifier with C = 5

		classical				metal		OVERALL	
	С	PRE	REC	f1	PRE	REC	f1	ACC	f1
	0.001	79.4	81.0	80.2	80.6	79.0	79.8	80.0	80.0
	0.010	77.2	78.0	77.6	77.8	77.0	77.4	77.5	77.5
	0.500	76.5	75.0	75.8	75.5	77.0	76.2	76.0	76.0
	1.000	88.2	75.0	81.1	78.3	90.0	83.7	82.5	82.4
	5.000	100.0	83.0	90.7	85.5	100.0	92.2	91.5	91.4
	10.000	100.0	78.0	87.6	82.0	100.0	90.1	89.0	88.9
	20.000	100.0	75.0	85.7	80.0	100.0	88.9	87.5	87.3
Confusion Matrix:									
	cla	met							
cla	41.50	8.50							
met	0.00	50.00							
Selected params: 5.00000									

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Predictions on Unseen testing data

```
data/music/test/classical.00095.au.wav:
P(classical=0.6365171558566964)
P(metal=0.3634828441433037)

data/music/test/metal.00004.au.wav:
P(classical=0.1743387880715576)
P(metal=0.8256612119284426)

data/music/test/rock.00037.au.wav:
P(classical=0.2757302369241449)
P(metal=0.7242697630758552)
```





Assumption: Unknown audio clips always belong to a single label

Real-world recordings are not segments of homogeneous content but sequences of segments of different labels

Audio Segmentation is about segmenting a long audio recording to a sequence of segments that are of homogenous content

- Supervised segmentation
- Unsupervised segmentation



Supervised



Supervised Audio Segmentation

Based on a pretrained segment model that can classify homogenous segments

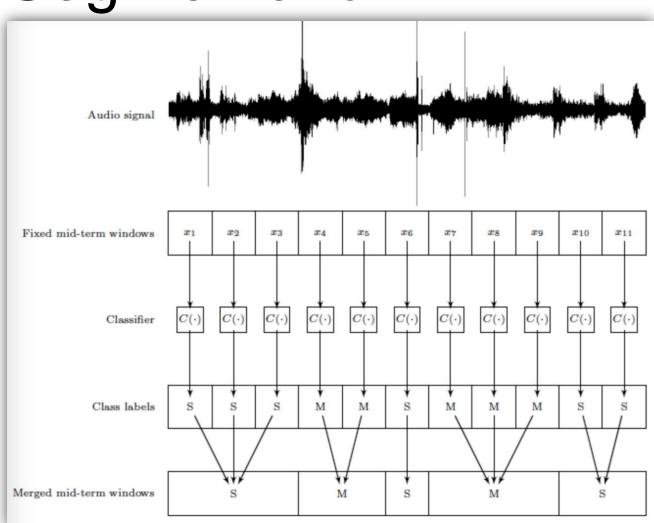
Pretrained segment model is needed



Supervised Audio Segmentation

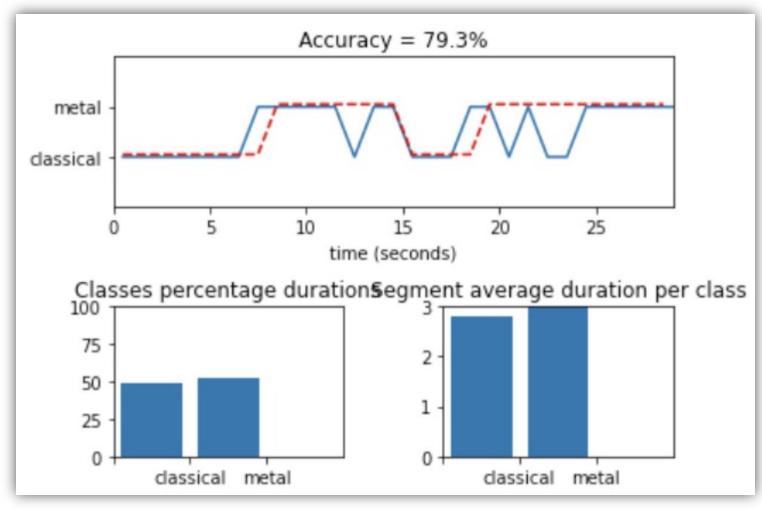
<u>Steps</u>

- Split audio clips into nonoverlapping fix-sized segments
- Classify each fix-sized segment using trained model
- 3. Merge successive segments that contain same class label





Supervised Audio Segmentation





Supervised Audio Segmentatio

Errors are mainly due to:

- Misclassifications of segment classifier
- 2. Time resolution issues

```
Segments:
segment 0 0.0 sec - 7.0 sec: classical
segment 1 7.0 sec - 12.0 sec: metal
segment 2 12.0 sec - 13.0 sec: classical
segment 3 13.0 sec - 15.0 sec: metal
segment 4 15.0 sec - 18.0 sec: classical
segment 5 18.0 sec - 20.0 sec: metal
segment 6 20.0 sec - 21.0 sec: classical
segment 7 21.0 sec - 22.0 sec: metal
segment 8 22.0 sec - 24.0 sec: classical
segment 9 24.0 sec - 29.0 sec: metal
```

```
Overall Accuracy: 0.79
Fix-sized segments:
fix-sized segment 0: classical
fix-sized segment 1: classical
fix-sized segment 2: classical
fix-sized segment 3: classical
fix-sized segment 4: classical
fix-sized segment 5: classical
fix-sized segment 6: classical
fix-sized segment 7: metal
fix-sized segment 8: metal
fix-sized segment 9: metal
fix-sized segment 10: metal
fix-sized segment 11: metal
fix-sized segment 12: classical
fix-sized segment 13: metal
fix-sized segment 14: metal
fix-sized segment 15: classical
fix-sized segment 16: classical
fix-sized segment 17: classical
fix-sized segment 18: metal
fix-sized segment 19: metal
fix-sized segment 20: classical
fix-sized segment 21: metal
fix-sized segment 22: classical
fix-sized segment 23: classical
fix-sized segment 24: metal
fix-sized segment 25: metal
fix-sized segment 26: metal
fix-sized segment 27: metal
fix-sized segment 28: metal
fix-sized segment 29: metal
```



Unsupervised



Unsupervised Audio Segmentation

Pretrained segment model is needed for supervised approach

In real-world, no data to train such a pretrained segment model is possible

Can you group song segments so that segments of the same group sound like they belong to the same song part?



Unsupervised Audio Segmentation

Grouping data points --> K-Means Clustering

Clusters of song segments may correspond to stuctural song elements if appropriate audio features are used



Unsupervised Audio Segmentation

Cluster 1



Cluster 2



Cluster 3



Cluster 4





Who speaks when?



Given a long audio sample with different speakers, group audio segments from a same speaker

Almost Unsupervised Audio Segmentation problem



Read the long audio file

```
input_file = "data/diarization_example.wav"
fs, x = read_audio_file(input_file)
```

Extract features



Create clusters on the extracted features

```
n_clusters = 4
x_clusters = [np.zeros((fs, )) for i in range(n_clusters)]
k_means = sklearn.cluster.KMeans(n_clusters=n_clusters)
k_means.fit(mt_feats_norm.T)
cls = k_means.labels_
```



Original Example



Hillary Clinton



Kamalla Harris



Donald Trump



Barrack Obama





Speech To Text (Speech Recognition)



Speech-To-Text (Speech Recognition)

Speech Recognition is a system that translates the language being spoken into digitized text format

<u>Features</u>

Streaming speech to text in real-time: the API is capable of processing real-time audio signals from the device microphone or take an audio file as input and convert it into text also.

Different models based on the domain: you can choose from different trained models depending on the requirements of the project. For example, for converting audio from a telephone, the enhanced phone call model can be used.

Adaptation: you can customize the API to understand rare words, currency, numbers etc by making these as additional classes.



Speech-To-Text (Speech Recognition)

Install the necessary library

```
pip install SpeechRecognition
```

Import the library

```
import speech_recognition as sr
```

Create the Speech-To-Text Recognizer object

```
recognizer = sr.Recognizer()
```



Speech-To-Text (Speech Recognition)

Listen to the microphone and convert speech to text

```
with sr.Microphone() as inputs:
    print("Please speak now")
    listening = recognizer.listen(inputs)
    print(recognizer.recognize_google(listening))
```

Listen to the audio file and convert speech to text

```
with sr.AudioFile('myvoice.wav') as inputs:
    file_audio = recognition.listen(inputs)
    try:
        convert_text = recognition.recognize_google(file_audio)
```



Text to Speech



Text to Speech

Google Text to Speech gTTS API

<u>Features</u>

- Customizable speech-specific sentence tokenizer that allows for unlimited lengths of text to be read, all while keeping proper intonation, abbreviations, decimals
- Customizable text pre-processors which can, for example, provide pronunciation corrections
- Automatic retrieval of supported languages



Text-To-Speech

Install the necessary library

```
pip install gTTS
```

Import the library

```
from gtts import gTTS
```

Create the Text-To-Speech Recognizer object

```
text='Hi ,Welcome to GBC AI Program'
language = 'en'
speech = gTTS(text = text, lang = language, slow = False)
```

Import the library

```
speech.save('medium_1.wav')
```



Further Reading

PyAudioAnalysis

https://github.com/tyiannak/pyAudioAnalysis

Feature Extraction

https://github.com/tyiannak/pyAudioAnalysis/wiki/3.-Feature-Extraction

https://github.com/tyiannak/basic_audio_analysis/blob/master/notebobok.ipynb

https://hackernoon.com/intro-to-audio-analysis-recognizing-sounds-using-machine-learning-qy2r3ufl



Further Reading

PyPDF2

https://pypi.org/project/PyPDF2/

Pyttsx3

https://pyttsx3.readthedocs.io/en/latest/