

# main

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## 1 Urban Sound Multiclass Classifier

### 1.0.1 ML1020 - Machine Learning at Scale - 2019

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### 1.0.2 Abstract:

The main objective of this project work is the case study of urban sounds audio event identification and classification. This is a supervised learning where we will be working on an audio event dataset with samples of audio data that belong to specific categories (which are the sources of the sounds).

We will be leveraging concepts from transfer learning and deep learning to build a robust classifier whereby, with any given audio sample belonging to one of our pre-determined categories, we should be able to correctly predict the source of this sound. The dataset we will be using is popularly known as the UrbanSound8K dataset. UrbanSounds8K dataset has 8,732 labeled audio sound files (the duration of which are usually equal to or greater than 4 seconds) that contain excerpts from common urban sounds.

### 1.0.3 Approach:

Document the steps we plan for our approach - either documentation or a diagram - need to finalize

## 1.1 Data Analysis

The dataset is downloaded from the following link: <https://urbansounddataset.weebly.com/urbansound8k.html>

Let's load the train data and explore available classes and sample distribution between the classes. note: the test data is not labeled thus it is useless for the training purpose. We have no choice but employ the train dataset to train and validate the model. The librosa module is an open source Python framework for audio and music analysis. We will be using this for analyzing as well as extracting features from audio data in subsequent sections.

```
In [ ]: # import required libraries
import pandas as pd
import numpy as np
import os
import matplotlib
```

```

import matplotlib.pyplot as plt
import seaborn as sns
import librosa
import librosa.display
import IPython.display
import soundfile as sf
import glob
import random

# define the path relative to the notebook source
# expected data structure
# -----
# project root
# src
#   main.ipynb
# data
#   original
#     train
#       samples
#       *.wav
#     train.csv
#     test
#       samples
#       *.wav
#     test.csv
TRAIN_DATA_PATH = "../data/original/train"
TRAIN_FILE = "train.csv"
# Read the train data classification info
classification = pd.read_csv("{0}/{1}".format(TRAIN_DATA_PATH, TRAIN_FILE), sep = ',')

```

In [89]: classification.head(10)

```

Out[89]:
   ID  Class
0   0  siren
1   1  street_music
2   2  drilling
3   3  siren
4   4  dog_bark
5   6  children_playing
6  10  street_music
7  11  drilling
8  12  gun_shot
9  15  dog_bark

```

In [90]: classification.info()

```

<class 'pandas.core.frame.DataFrame'>
RangeIndex: 5435 entries, 0 to 5434
Data columns (total 2 columns):

```

```
ID          5435 non-null int64
Class       5435 non-null object
dtypes: int64(1), object(1)
memory usage: 85.0+ KB
```

Let's examine what classes are available

```
In [91]: classes = classification.Class.unique()
         classes
```

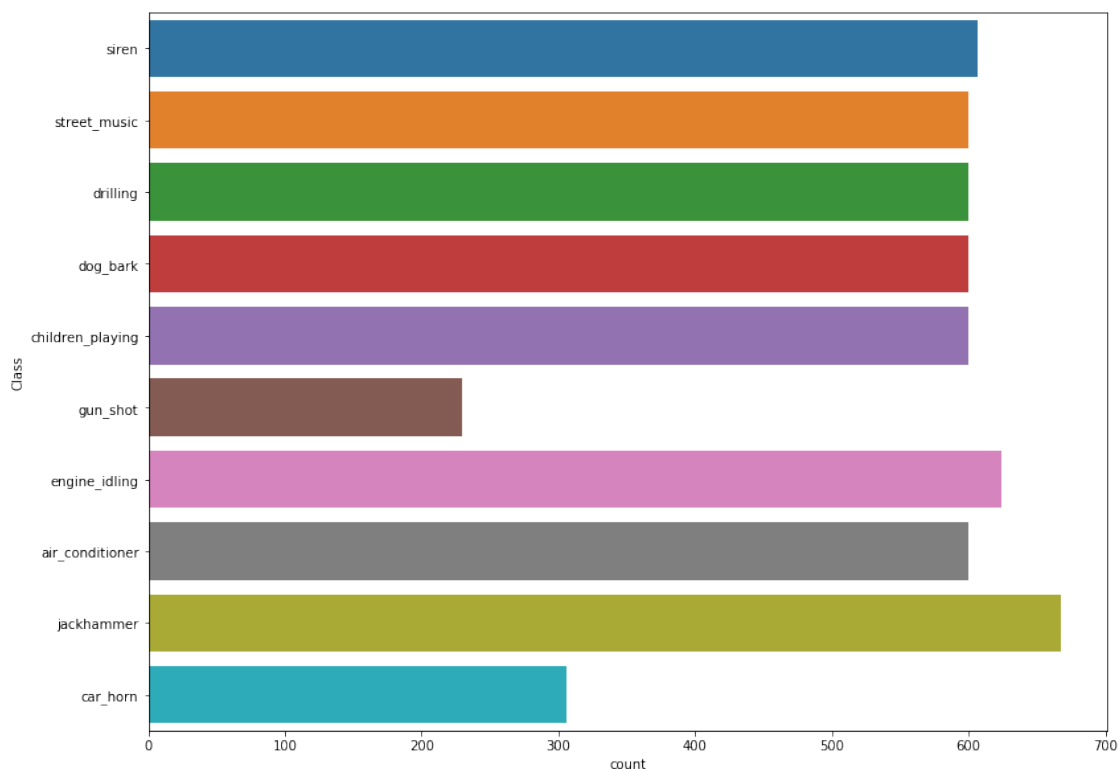
```
Out[91]: array(['siren', 'street_music', 'drilling', 'dog_bark',
                'children_playing', 'gun_shot', 'engine_idling', 'air_conditioner',
                'jackhammer', 'car_horn'], dtype=object)
```

```
In [92]: print("Number of classes: {}".format(len(classes)))
```

Number of classes: 10

As per the output above the dataset comprises ten urban sound classes. It is time to take a look at the class population.

```
In [93]: plt.figure(figsize=(13,10))
         p = sns.countplot(data=classification,y='Class')
```



Gun Shot and Car Horn categories are underpopulated. We could:

- digitize and upsample these categories
- downsample all categories
- add more labeled observations to the smaller categories
- leave as is hoping that the categories with the smaller population still have enough to train the network

Let's leave the data intact. If during the training we realize that the sparsely populated categories are not detected well we will take action

### 1.1.1 Basic Data Stats

Let's explore the sound samples in depth. Knowing the sample duration, sample, rate, number of channels, etc. will help us to understand what the data normalization and preparation must be done prior to feeding the data to a model. We start with the extraction of sound characteristics of each sample...

```
In [94]: def classEncoder(className):
        classMap = {
            'siren':1,
            'street_music':2,
            'drilling':3,
            'dog_bark':4,
            'children_playing':5,
            'gun_shot':6,
            'engine_idling':7,
            'air_conditioner':8,
            'jackhammer':9,
            'car_horn':10
        }
        return classMap.get(className)

In [96]: # grab the full paths of all sample files
paths = glob.glob("{0}/samples/*".format(TRAIN_DATA_PATH))
# create a dataframe
sounds = []
for path in paths:
    fn = int(os.path.splitext(os.path.basename(path))[0])
    category = classification[classification.ID == fn]
    audio = sf.SoundFile(path)
    cls = category.Class.values[0]
    sounds.append((fn, classEncoder(cls), cls, audio.name, audio.channels, audio.sample_rate,
        round(audio.frames/audio.samplerate,1), audio.subtype))
df = pd.DataFrame(sounds, columns=['ID', 'ClassID', 'Class', 'Path', 'Channels', 'SampleRate', 'Duration', 'Subtype'])
df.head(20)
```

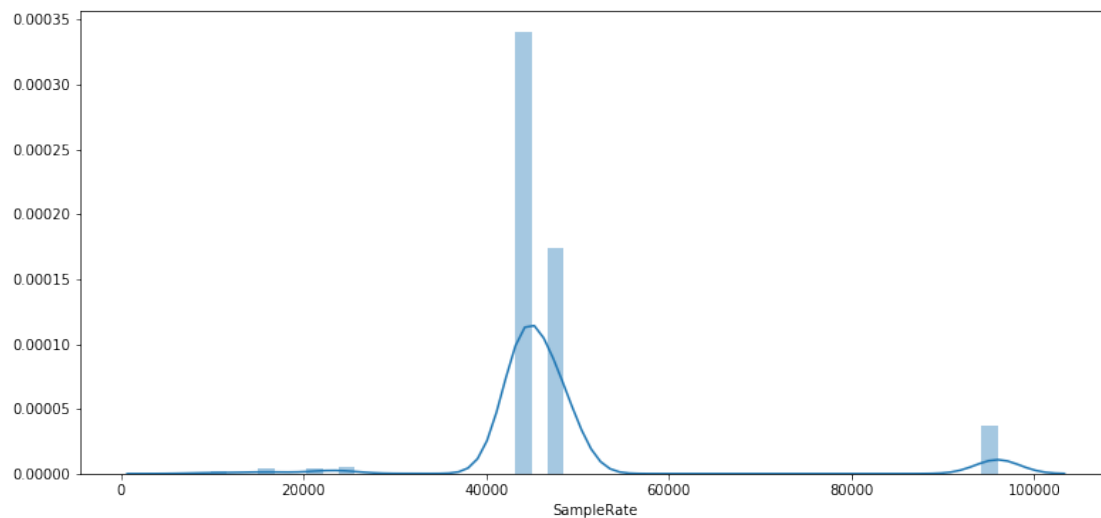
```
Out [96]:
```

	ID	ClassID	Class	Path \
0	0	1	siren	../data/original/train/samples\0.wav
1	1	2	street_music	../data/original/train/samples\1.wav
2	10	2	street_music	../data/original/train/samples\10.wav
3	100	8	air_conditioner	../data/original/train/samples\100.wav
4	1000	2	street_music	../data/original/train/samples\1000.wav
5	1001	8	air_conditioner	../data/original/train/samples\1001.wav
6	1003	4	dog_bark	../data/original/train/samples\1003.wav
7	1004	8	air_conditioner	../data/original/train/samples\1004.wav
8	1006	9	jackhammer	../data/original/train/samples\1006.wav
9	1007	10	car_horn	../data/original/train/samples\1007.wav
10	1008	10	car_horn	../data/original/train/samples\1008.wav
11	101	3	drilling	../data/original/train/samples\101.wav
12	1014	5	children_playing	../data/original/train/samples\1014.wav
13	1015	2	street_music	../data/original/train/samples\1015.wav
14	1017	1	siren	../data/original/train/samples\1017.wav
15	1018	2	street_music	../data/original/train/samples\1018.wav
16	1021	4	dog_bark	../data/original/train/samples\1021.wav
17	1022	5	children_playing	../data/original/train/samples\1022.wav
18	1024	5	children_playing	../data/original/train/samples\1024.wav
19	1025	9	jackhammer	../data/original/train/samples\1025.wav

	Channels	SampleRate	Frames	Format	Length	SubType
0	2	44100	176400	WAV	4.0	PCM_16
1	1	48000	192000	WAV	4.0	PCM_16
2	2	44100	176400	WAV	4.0	PCM_16
3	2	44100	176400	WAV	4.0	PCM_16
4	2	44100	176400	WAV	4.0	PCM_16
5	2	44100	176400	WAV	4.0	PCM_16
6	2	96000	384000	WAVEX	4.0	PCM_24
7	2	44100	176400	WAV	4.0	PCM_16
8	2	44100	176400	WAV	4.0	PCM_16
9	1	16000	64000	WAV	4.0	PCM_16
10	2	48000	17561	WAV	0.4	PCM_16
11	2	48000	55200	WAVEX	1.1	PCM_24
12	2	44100	55125	WAV	1.2	PCM_16
13	2	48000	192000	WAV	4.0	PCM_16
14	2	48000	192000	WAVEX	4.0	PCM_24
15	2	44100	176400	WAV	4.0	PCM_16
16	1	48000	192000	WAVEX	4.0	PCM_24
17	2	44100	176400	WAV	4.0	PCM_16
18	2	48000	192000	WAVEX	4.0	PCM_24
19	2	44100	176400	WAV	4.0	PCM_16

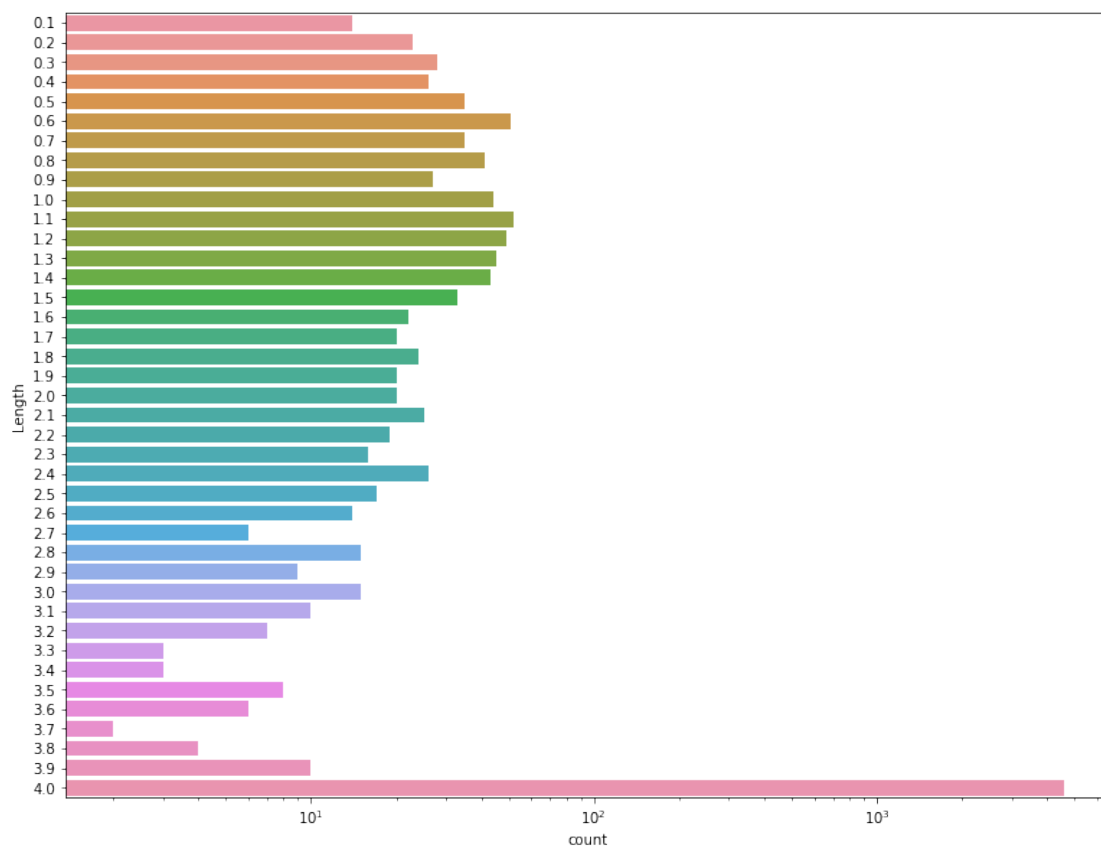
## Sample Rate Distirbution

```
In [97]: plt.figure(figsize=(13,6))
sns.distplot(df.SampleRate);
```



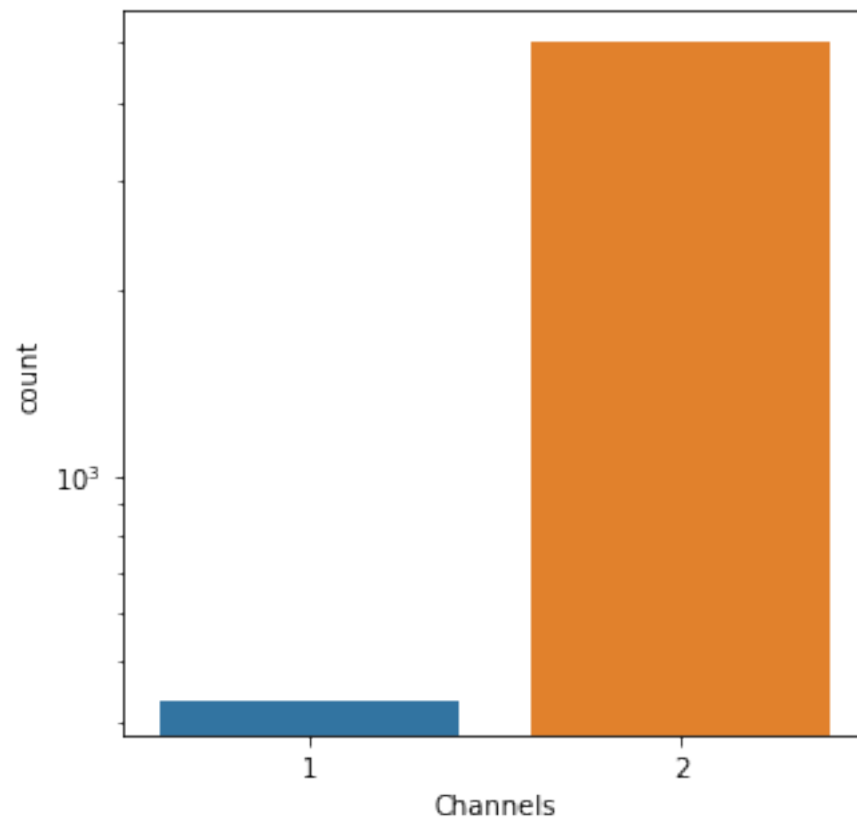
Evidently the sample rate of the sound files varies. We would have to resample the original data to bring it to the same standard. Let's calculate the length of the sounds

```
In [98]: plt.figure(figsize=(13,10))
         sns.countplot(y=df.Length, log=True);
```



Majority of the sound files are 4 second long. But there are some file that are less than a second long. Designing the model we would have to make sure that the input layer is able to deal with the sound samples of various length and a sample rate. We might also filter out the samples that less than 0.5 second long, because not likely they do not carry to valuable information. Lastly we are going to verify how many channels the recorded audio file have (stereo vs mono)

```
In [99]: plt.figure(figsize=(5,5))
sns.countplot(x=df.Channels, log=True);
```



Just a few audio files were recorded in mono. For model training we probably would need just one channel (TBD)

**Sound Characteristics of Each Category** There are ten sound classes. Let's take a sample from each class and review its characteristics. **Note:** we will be using librosa library to visualize the features of the audio sample.

```
In [100]: sampleDf=df.groupby('Class',as_index = False,group_keys=False).apply(lambda s: s.sample(
sampleDf
```

```

Out[100]:
      ID  ClassID      Class \
1996  3895        8  air_conditioner
2950  5299       10    car_horn
590   189        5  children_playing
874   2303        4    dog_bark
4579  7626        3    drilling
1837  3670        7  engine_idling
4034  6872        6    gun_shot
3605  6263        9  jackhammer
1911  3771        1    siren
1813  3644        2  street_music

      Path  Channels  SampleRate  Frames \
1996  ../data/original/train/samples\3895.wav      2      48000  192000
2950  ../data/original/train/samples\5299.wav      2      48000   60960
590   ../data/original/train/samples\189.wav       2      48000  192000
874   ../data/original/train/samples\2303.wav      1      22050   88200
4579  ../data/original/train/samples\7626.wav      2      44100  176400
1837  ../data/original/train/samples\3670.wav      2      44100  176400
4034  ../data/original/train/samples\6872.wav      2      48000  100320
3605  ../data/original/train/samples\6263.wav      2      96000  384000
1911  ../data/original/train/samples\3771.wav      1      44100  176400
1813  ../data/original/train/samples\3644.wav      2      24000   96000

      Format  Length  SubType
1996  WAVEX     4.0  PCM_24
2950   WAV     1.3  PCM_16
590   WAV     4.0  PCM_16
874   WAV     4.0  PCM_16
4579  WAV     4.0  PCM_16
1837  WAV     4.0  PCM_16
4034  WAV     2.1  PCM_16
3605  WAVEX     4.0  PCM_24
1911  WAV     4.0  PCM_16
1813  WAV     4.0  PCM_16

```

Out of curiosity let's listen a few sounds from our sample collection

### Drilling

```
In [101]: IPython.display.Audio(data=sampleDf[sampleDf.Class == 'drilling'].Path.values[0])
```

```
Out[101]: <IPython.lib.display.Audio object>
```

### Children Playing

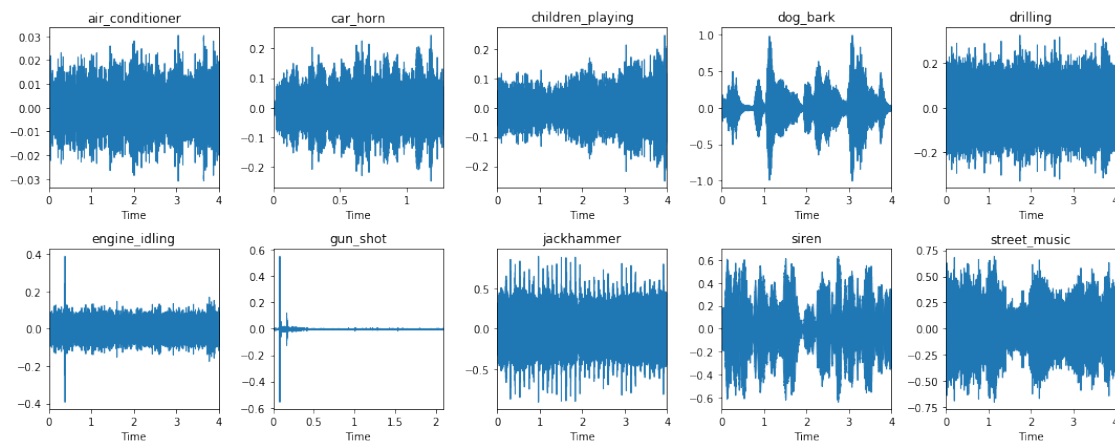
```
In [102]: IPython.display.Audio(data=sampleDf[sampleDf.Class == 'children_playing'].Path.values[0])
```

```
Out[102]: <IPython.lib.display.Audio object>
```



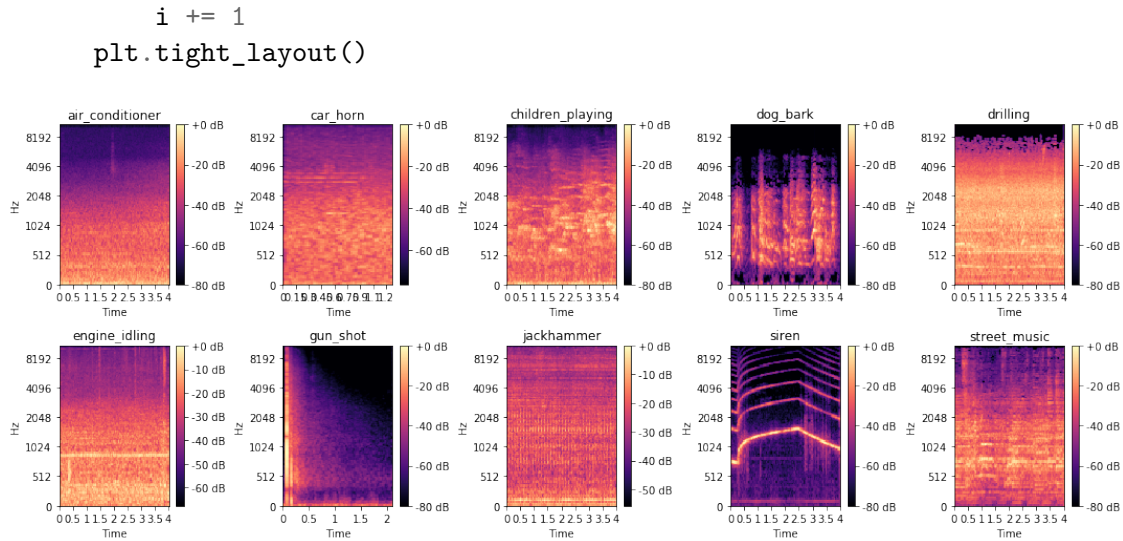
**Visualize Data** Let's now visualize what these different audio sources look like by plotting their waveforms. This will be a waveform amplitude plot for each audio sample:

```
In [103]: audio=[sf.read(path) for path in sampleDf["Path"]]
i = 0
fig = plt.figure(figsize=(15, 6))
for sample in audio:
    plt.subplot(2, 5, i+1)
    y = librosa.resample(sample[0].T, sample[1], 22050)
    y = librosa.to_mono(y)
    librosa.display.waveplot(y)
    plt.title(sampleDf["Class"].iloc[[i]].values[0])
    i += 1
plt.tight_layout()
```



The waveform charts rendered above clearly show that each sound class has distinctive characteristics, such as rhythm, amplitude, etc. Though some classes are more similar than others, for example Air Conditioner and Engine Idling feature rather similar, monotonous sound. Let's apply another popular technique that exposes the features of the sound even better **mel spectrogram**. The name mel comes from the word melody. This indicates that the scale is based on pitch comparisons. The melscale is thus a perceptual scale of pitches that have been judged by listeners to be equal in distance from one another.

```
In [104]: i = 0
fig = plt.figure(figsize=(15, 6))
for sample in audio:
    plt.subplot(2, 5, i+1)
    y = librosa.resample(sample[0].T, sample[1], 22050)
    y = librosa.to_mono(y)
    M = librosa.feature.melspectrogram(y=y)
    librosa.display.specshow(librosa.power_to_db(M, ref=np.max), y_axis='mel', x_axis='time')
    plt.title(sampleDf["Class"].iloc[[i]].values[0])
    plt.colorbar(format='%+02.0f dB')
```



Evidently the mel spectrograms have more features than the amplitude/time waveform presentation. Now the distinction between the Engine Idling and Air Conditioner is much clearer. Another advantage of such data presentation is that it could be fed to the **Convolutional Neural Network**.

## 1.2 Feature Engineering

To be continued...

In [ ]:

### 1.2.1 Model development:

Document the steps for Transfer Learning

**AWS or GCP setup/approach:** We need to outline the approach we plan to take in either GCP or AWS for handling big dataset in cloud.

**Conclusion:** TODO

**References:** Book - Hands on Transfer Learning With Python - PACKT Publishing  
Download Link: <https://urbansounddataset.weebly.com/urbansound8k.html>

In [ ]:

In [ ]: