dataset_exploration

July 3, 2020

0.1 LibriSpeech Structure

The LibriSpeech corpus is a standard benchmark for recent speech related neural network and machine learning research << CITE >>. It contains various sets of speech samples (urls) of readers reading from texts contained within the Project Gutenberg corpus of books. The speech samples are saved within speaker specific directories in the .flac file format. The follow urls are contained within it: * dev-clean * dev-other * test-clean * test-other * train-clean-100 (100 hours) * train-clean-360 (360 hours) * train-other-500 (500 hours)

All of the audio has a sampling rate of 16kHz.

The LibriSpeech dataset (thankfully) has a standardized structure of its directories:

This makes it useful for most speech-related tasks, be it speech recognition or speaker recognition / verfication.

Let's investigate on sample file in particular to just to get a sense of how we would interact with it.

0.2 Audio Sample

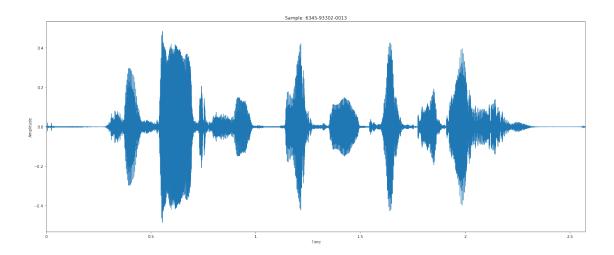
In [63]: librispeech_sr = 16000

Here we'll load and display the audio file using the librosa library which provides many helpful utilities for dealing with audio files. Something worth noting here is that the default sample rate for librosa is typically 22050Hz.

Now given that raw audio in waveform, we can visualize the amplitude (?) of the wave over the time of the sample.

y, _ = librosa.core.load(sample_file_path, sr=librispeech_sr)

```
In [80]: # TODO: add axis label
    plt.figure(figsize=(25,10))
    plt.title(f'Sample: {sample}')
    plt.ylabel('Amplitude')
    _ = librosa.display.waveplot(y, sr=librispeech_sr)
```



What does the sample actually sound like?

```
In [68]: IPython.display.Audio(y, rate=librispeech_sr)
Out[68]: <IPython.lib.display.Audio object>
```

And, as mentioned, LibriSpeech also provides the transcript of the audio file, which will come in handy for performing speech recognition.

6345-93302-0013 SHE SAID HOW FRIGHTFULLY COLD IT IS