

SPEECH PROCESSING ASSIGNMENT - 2

GitHub Link : <https://github.com/Toji339/SpeechProcessing>

Dataset Description :

- This LJ Speech dataset has been taken from the Kaggle.
- It contains 13,100 short audio clips of a single speaker
- These clips vary in length from 1 to 10 sec and have a total length of around 24 hours.

Objective :

- ▶ Load and preprocess a speech signal (convert to mono and resample to 16kHz).
- ▶ Use a pre-trained Wav2Vec2 model to recognize phonemes from the speech signal.
- ▶ Extract specific phoneme segments based on their time intervals.
- ▶ Visualize the phoneme waveforms and label them according to the recognized phonemes.

Code :

```
1
2  import torchaudio
3  import torchaudio.transforms as T
4  import librosa.display
5  import torch
6  from transformers import Wav2Vec2Processor, Wav2Vec2ForCTC
7  import matplotlib.pyplot as plt
8  import numpy as np
9  import nltk
10 from nltk.corpus import cmudict
11
12 nltk.download('cmudict')
13 cmu_dict = cmudict.dict()
14
15 audio_path = "/content/LJ001-0030.wav"
16 waveform, sample_rate = torchaudio.load(audio_path)
17
18 resampler = T.Resample(orig_freq=sample_rate, new_freq=16000)
19 waveform = resampler(waveform.mean(dim=0, keepdim=True))
20
21 processor = Wav2Vec2Processor.from_pretrained("facebook/wav2vec2-large-960h")
22 model = Wav2Vec2ForCTC.from_pretrained("facebook/wav2vec2-large-960h")
23
```

```

24 input_values = processor(waveform.squeeze().numpy(), return_tensors="pt",
25 sampling_rate=16000).input_values
26 with torch.no_grad():
27     logits = model(input_values).logits
28
29 predicted_ids = torch.argmax(logits, dim=-1)
30 transcription = processor.batch_decode(predicted_ids)[0].upper()
31 print("Recognized Text:", transcription)
32
33 words = transcription.split()
34 phonemes = []
35 for word in words:
36     if word in cmu_dict:
37         phonemes.extend(cmu_dict[word][0])
38     else:
39         phonemes.append(word)
40
41 print("Recognized Phonemes:", " ".join(phonemes))
42
43 total_duration = waveform.shape[1] / 16000
44 time_intervals = np.linspace(0, total_duration, num=len(phonemes))
45
46 plt.figure(figsize=(12, 4))
47 librosa.display.waveshow(waveform.squeeze().numpy(), sr=16000)
48 plt.xlabel("Time (s)")
49 plt.ylabel("Amplitude")
50 plt.title("Phoneme Visualization")
51
52
53 for i, phoneme in enumerate(phonemes):
54     if i < len(time_intervals):
55         plt.text(time_intervals[i], 0, phoneme, fontsize=10, ha='center',
56                 color='red', rotation=45)
57
58 plt.show()

```

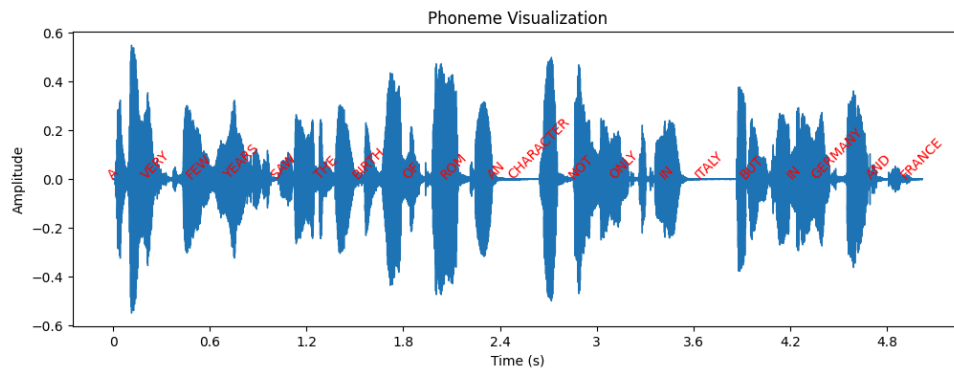


Figure 1: Final Plot with recognized text .