

# Programming Assignment 01

## Report

Renu Sankhla (B21AI028)

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### 1 Task 0

In this task i take two sentence write below and upload it's hindi and english version.

**Sentence1** = "What you think of yourself matters much more than what others think about you. You should think wisely."

**Sentence2** = "There is no need of any competition with anybody. You are yourself and as you are, you are perfectly good. Accept yourself."



Figure 1: An image of audio's

### 2 Task A :

In this task we have to utilizes a pre-trained Massively Multilingual Speech (MMS) Language Identification (LID) model to identify languages from audio

files. So we first load the audio files containing both English and Hindi speech. Then, we process each audio sample using the pre-trained model to obtain language predictions. Finally, we print out the detected language for each audio sample. The model's performance in identifying languages can be evaluated by comparing the predicted languages with the ground-truth languages of the audio files. We find that the model has 100 percent accurate result for our

```
detected language is hin
detected language is hin
detected language is eng
detected language is eng
```

Figure 2: An image of audio's

input audio's.

### 3 Task B :

In this task we use the pre-trained Massively Multilingual Speech (MMS) Text-to-Speech (TTS) models like "facebook/mms-tts-hin" to generate speech for Hindi language sentences and we use the "facebook/mms-tts-eng" model for English sentence. So first we take two sentences in Hindi and English and then we create a tokenizer and train these tokens on our model and generate the audio.

```
input = tokenizer(text = "What you think of yourself matters much more than what others think about you. You should think wisely.", return_tensors = "pt")
set_seed(555)
with torch.no_grad():
    outputs = model(**input)

waveform = outputs.waveform[0]
Audio(waveform, rate = 16000)
```

Figure 3: An image of text to speech generation for English language text

```
input = tokenizer(text = "आप क्या सोचते हैं अपने बारे में ज्यादा मानने रखता है इसलिए कि दूसरे क्या सोचते हैं आपके बारे में", return_tensors = "pt")
set_seed(555)
with torch.no_grad():
    outputs = model(**input)

waveform = outputs.waveform[0]
Audio(waveform, rate = 16000)
```

Figure 4: An image of text to speech generation for Hindi language text

## 4 Task C :

In this task we first loads a pre-trained Massively Multilingual Speech (MMS) model, utilizing an automatic processor for feature extraction, followed by a Wav2Vec2 model adapted for connectionist temporal classification (CTC) task, using this we generate the transcription of our english and hindi audio files.

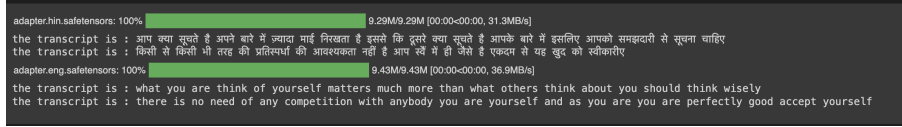


Figure 5: An image of generated transcript of audio using ASR

than we calculate the Character Error Rate (CER) and Word Error Rate (WER) for the Automatic Speech Recognition (ASR) transcriptions compared to the ground truth transcriptions for both Hindi and English languages. CER measures the percentage of characters that are incorrectly recognized by the ASR system compared to the ground truth. WER measures the percentage of words that are incorrectly recognized. Lower CER and WER values indicate better performance of the ASR system.

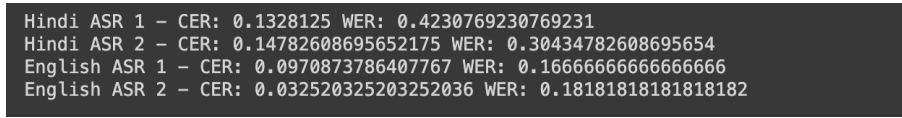


Figure 6: An image of generated transcript of audio using ASR

We can see these scores that the ASR system performs better on English transcriptions compared to Hindi transcriptions. Additionally, we can see that from ASR 2 generally outperforms ASR 1 in terms of both CER and WER for both languages. The differences in scores could be due to factors such as language complexity, pronunciation variations, and the quality of the ASR model used.