

Malayalam Text-to Speech system

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Objective

- To build a Text to Speech(TTS) system in Malayalam
- Obtain the state of art result

- Module1 : EDA, dataset collection
- Module2: Train first TTS system in Malayalam
- Module3: Fine tune TTS system
- Module4: User Interface

Work Done

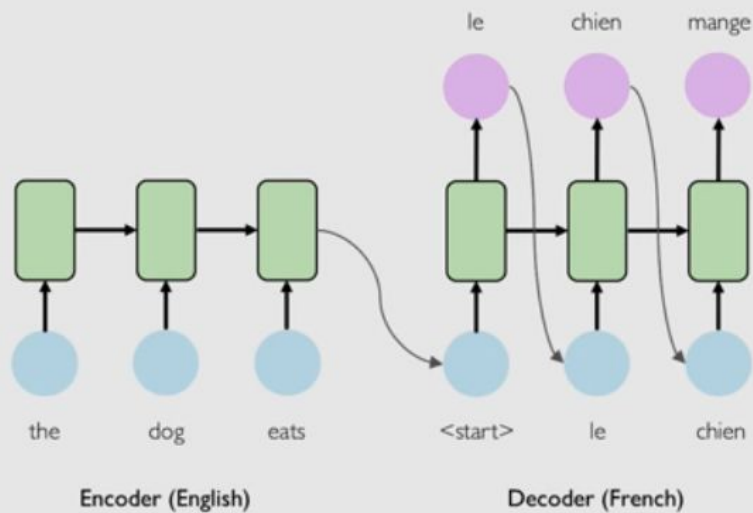
Tactron paper Implementation

Tactron model is a sequence to sequence model, which is Currently one of the best Text to speech architectures. An improved architecture called Tactron2 also exists. Seq-2-seq models use RNNs as backbone

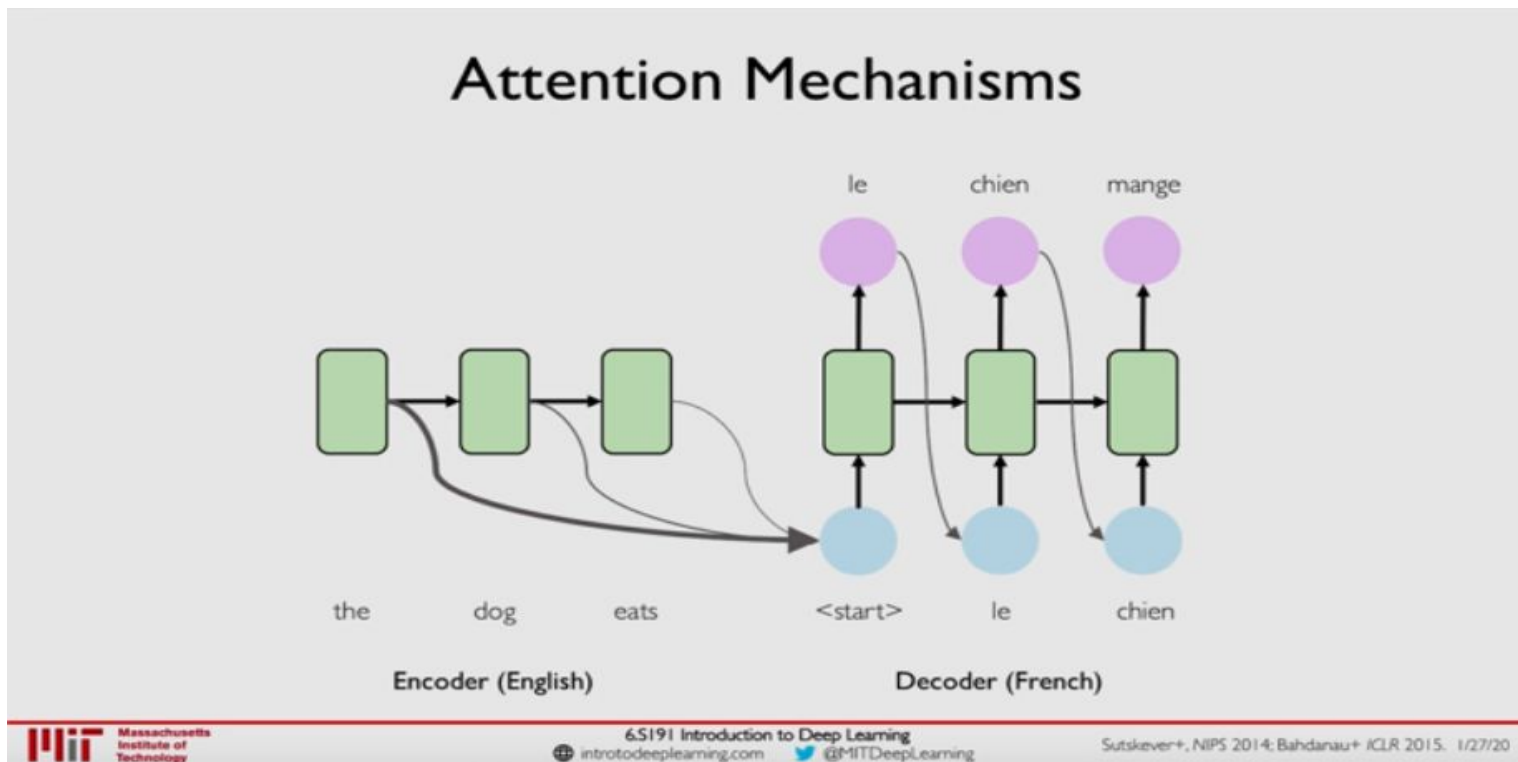
Aim: To understand the paper and implement architecture

Sequence to Sequence models

Example Task: Machine Translation



Sequence to Sequence models



Components of Tactron architecture

1. CBHG module
2. Encoder
3. Decoder

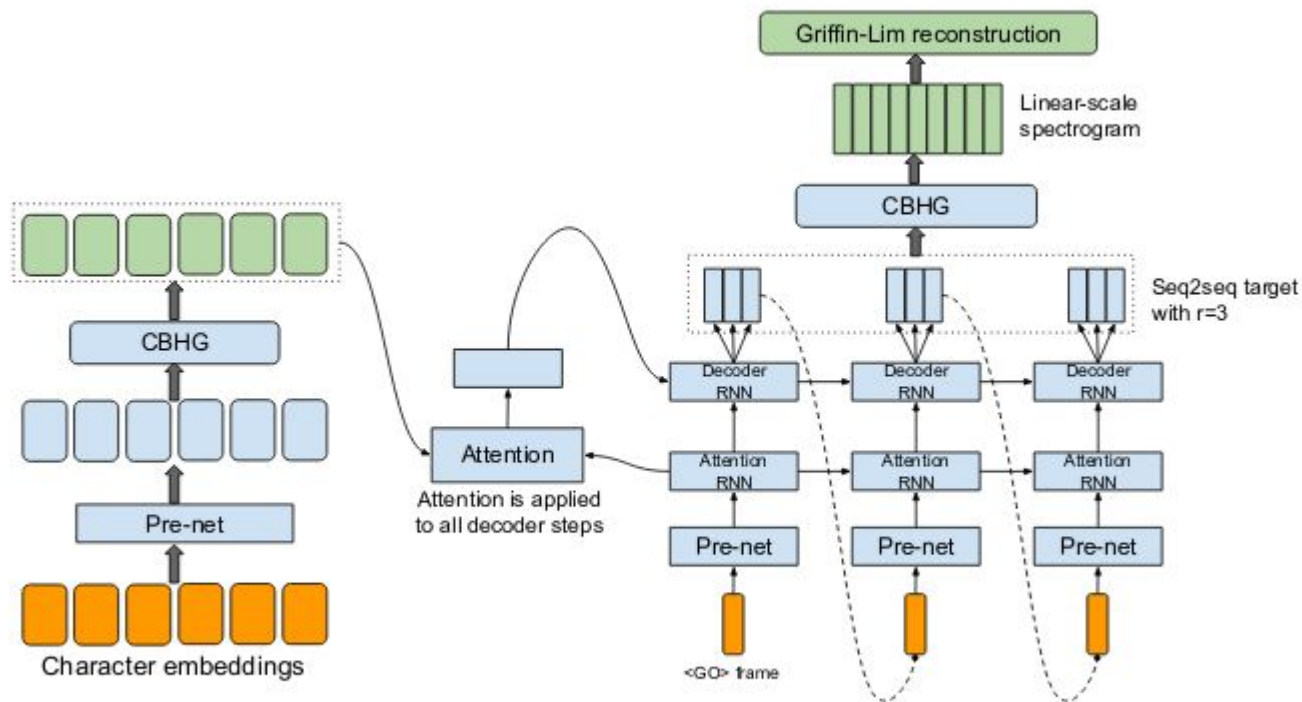


Figure 1: *Model architecture. The model takes characters as input and outputs the corresponding raw spectrogram, which is then fed to the Griffin-Lim reconstruction algorithm to synthesize speech.*

CBHG Module

```
In [4]: class CBHG(nn.Module):
        def __init__(self, in_f, K=16, conv_bank_features=128, conv_projections=[128, 128], hig
            hway_features=128,
                gru_features=128, num_highways=4):
            super(CBHG, self).__init__()
            self.in_f = in_f
            self.conv_b_f = conv_b_f
            self.hf = hf
            self.gru_f = gru_f
            self.relu = nn.ReLU()

            self.convl_d_banks = nn.ModuleList([
                BatchNormConvD(in_f, conv_bank_features, kernel_size=k, stride=1, padding=[(k
- 1) // 2, k // 2], activation=self.relu) for k in range(1, K+1)
            ])

            out_f = [K * conv_bank_features] + conv_projections[:-1]
            activations = [self.relu] * (len(conv_projections) - 1)
            activations += [None]

            layer_set = []
            for (in_s, out_s, ac) in zip(out_f, conv_projections[:-1], activations):
                layer = BatchNormConvD(in_size,
                                        out_size,
                                        kernel_size=3,
                                        stride=1,
                                        padding=[1, 1],
                                        activation=ac)

                layer_set.append(layer)

            self.convl_d_projections = nn.ModuleList(layer_set)
            # setup Highway layers
            if self.highway_features != conv_projections[-1]:
                self.pre_highway = nn.Linear(conv_projections[-1],
                                                highway_features,
                                                bias=False)

            self.highways = nn.ModuleList([
                Highway(highway_features, highway_features) for _ in range(num_highways)
            ])
            self.gru = nn.GRU(gru_features, gru_features, 1,
                              batch_first=True, bidirectional=True)

        def forward(self, inputs):
            x = inputs
            outs = []
            for convld in convld_banks:
                out = convld(x)
                outs.append(out)
            x = torch.cat(outs, dim=1)
            assert x.size(1) == self.conv_bank_features * len(self.convl_d_banks)
            for convld in self.convl_d_projections:
                x = convld(x)
            x += inputs
            x = x.transpose(1, 2)
            if self.highway_features != self.conv_projections[-1]:
                x = self.pre_highway(x)

            for highway in self.highways:
                x = highway(x)
            self.gru.flatten_parameters()
            outputs, _ = self.gru(x)
            return outputs
```

Encoder Module

```
In [11]: class Prenet(nn.Module):
    def __init__(self, in_f, pre_dropout=True, out_f=[256, 256], bias=True):
        super(Prenet, self).__init__()
        self.pre_dropout = pre_dropout
        # excluding output feature of last layer
        in_f = in_f + out_f[:-1]
        self.layers = nn.ModuleList([
            Linear(in_size, out_size, bias=bias)
            for in_size, out_size in zip(in_f, out_f)
        ])

    def forward(self, x):
        for linear in self.layers:
            if self.pre_dropout:
                F.dropout(F.relu(linear(x)), p=0.5, train=self.training)
            else:
                F.relu(linear(x))
        return x
```

```
In [12]: class Encoder(nn.Module):
    """Encapsulate Prenet and CBHG modules for encoder"""

    def __init__(self, in_features):
        super(Encoder, self).__init__()
        self.prenet = Prenet(in_features, out_features=[256, 128])
        self.cbhg = EncoderCBHG()

    def forward(self, inputs):
        # B x T x prenet_dim
        outputs = self.prenet(inputs)
        outputs = self.cbhg(outputs.transpose(1, 2))
        return outputs
```

Text normalisation

```
[ ] from indicnlp.normalize.indic_normalize import IndicNormalizerFactory
input_text="""ഹെ.സി.സി പെരുമാറ്റച്ചട്ടം 2.22 വകുപ്പ് അനുസരിച്ചാണ്"""
remove_nuktas=True
factory=IndicNormalizerFactory()
normalizer=factory.get_normalizer("mal")
print(normalizer.normalize(input_text))
```

☞ ഹെ.സി.സി പെരുമാറ്റച്ചട്ടം 2.22 വകുപ്പ് അനുസരിച്ചാണ്

```
[ ] from indicnlp.tokenize import sentence_tokenize

indic_string="""ഹെ.സി.സി പെരുമാറ്റച്ചട്ടം 2.22 വകുപ്പ് അനുസരിച്ചാണ്"""

# Split the sentence, language code "hi" is passed for hingi
sentences=sentence_tokenize.sentence_split(indic_string, lang='ml')

# print the sentences
for t in sentences:
```

Work on top of Mozilla TTS to train first notebook

- Collect data
- Do dataset preprocessing
- Text normalisation
- Initial training pipeline

Training Notebook

<https://drive.google.com/drive/u/2/folders/1gZbGSVbVnaX87XO9E-nGYxJcZC4f12rg>

EDA of Malayalam Speech corpora

Total duration of dataset

```
In [4]: sum=0
        for dirname, _, filenames in os.walk('../data/msc-master/audio/'):
            for filename in filenames:
                y, sr = librosa.load(os.path.join(dirname, filename))
                sum = sum + librosa.get_duration(y=y,sr=sr)

/home/kurian/data/.env/lib/python3.7/site-packages/librosa/core/audio.py:161: UserWarning: PySoundFile failed. Trying
audioread instead.
  warnings.warn('PySoundFile failed. Trying audioread instead.')
/home/kurian/data/.env/lib/python3.7/site-packages/librosa/core/audio.py:161: UserWarning: PySoundFile failed. Trying
audioread instead.
  warnings.warn('PySoundFile failed. Trying audioread instead.')
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audioread instead.
  warnings.warn('PySoundFile failed. Trying audioread instead.')

In [5]: print(f'Total duration of Audio sample in MSC data: {sum/60} minutes')

Total duration of Audio sample in MSC data: 53.91714814814814 minutes
```

Research paper

A study on Text to speech systems for Non-English languages

(featured in IJRAR Research Journal)

<http://www.ijrar.org/papers/IJRAR19K8100.pdf>

Thank you!