

Voice Cloning with Tacotron2



Creating a voice cloning system using deep learning involves several steps, including data preparation, model training, and synthesis. Here, I'll provide a simplified Python code outline using the Tacotron2 and WaveGlow models from NVIDIA, which are commonly used for text-to-speech (TTS) applications.

Prerequisites

1. **Install Dependencies**: Make sure you have Python installed, along with PyTorch, and NVIDIA's Tacotron2 and WaveGlow repositories.

```
pip install torch
pip install numpy scipy
git clone https://github.com/NVIDIA/tacotron2.git
cd tacotron2
pip install -r requirements.txt
git clone https://github.com/NVIDIA/waveglow.git
cd waveglow
pip install -r requirements.txt
```

Step 1: Data Preparation

You need a dataset consisting of audio files and their corresponding transcripts. For simplicity, we'll assume you have a dataset in the format required by Tacotron2.

Step 2: Preprocess the Data

Preprocessing involves converting audio files into mel-spectrograms and normalizing the transcripts.

Step 3: Train Tacotron2 Model

Train the Tacotron2 model on the preprocessed data.

```
import torch
from tacotron2.model import Tacotron2
from tacotron2.train import train

# Initialize the model
model = Tacotron2()

# Load the dataset
train_loader = ...

# Train the model
train(model, train_loader)
```

Step 4: Synthesize Speech with Tacotron2 and WaveGlow

Once the Tacotron2 model is trained, use it to generate mel-spectrograms from text, and then use WaveGlow to convert mel-spectrograms into audio.

```
python
  import torch
  from tacotron2.model import Tacotron2
  from tacotron2.text import text_to_sequence
  from waveglow.denoiser import Denoiser
  # Load Tacotron2 and WaveGlow models
 tacotron2 = Tacotron2()
waveglow = torch.load('path_to_waveglow_model')
  denoiser = Denoiser(waveglow)
  # Function to generate speech
  def generate_speech(text, tacotron2, waveglow, denoiser):
      sequence = np.array(text_to_sequence(text, ['english_cleaners']))[None, :]
      sequence = torch.from_numpy(sequence).to(device='cuda', dtype=torch.long)
      with torch.no_grad():
           mel_outputs, mel_outputs_postnet, _, alignments = tacotron2.inference(sequence)
audio = waveglow.infer(mel_outputs_postnet, sigma=0.666)
audio = denoiser(audio, strength=0.01)[:, 0]
      return audio.cpu().numpy()
  # Generate speech
  text = "Hello, how are you?"
  audio = generate_speech(text, tacotron2, waveglow, denoiser)
  # Save to file
  import soundfile as sf
  sf.write('output.wav', audio, 22050)
```

Notes

1. **Data Quality**: The quality of the cloned voice depends heavily on the quality and quantity of your training data.



- 2. **Training Time**: Training a Tacotron2 model from scratch can take a significant amount of time and computational resources. Consider using pre-trained models if available.
- 3. **Fine-Tuning**: If you have a pre-trained model, you can fine-tune it on your specific dataset to adapt it to your voice cloning task.

This outline provides a basic framework. For a production-level system, you'd need to handle more aspects such as model checkpoints, data augmentation, error handling, and more robust preprocessing.

ChatGPT can make mistakes. Check important info.