Guru Nanak Dev Engineering College

Training Diary - TR-102 Report

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Day 7

Training Summary

On the seventh day of training, we explored the complete cycle of voice-based Al systems by building:

- A Text-to-Speech (TTS) converter using xtts, and
- A Speech-to-Speech (S2S) converter by combining Whisper and xtts.

These applications allow machines to understand human speech and respond audibly, forming the foundation for voice assistants, dubbing systems, and accessibility tools.

Project 1: Text-to-Speech Converter (TTS)

We first created a Text-to-Speech tool using **xtts (Extended Text-to-Speech)**, which converts written input into spoken audio.

Steps Involved:

- Accept **text input** from the user.
- Process the text using xtts, a neural network trained for lifelike speech synthesis.
- Generate and play the **speech output**.
- Save the audio as a file (e.g., .wav format) for playback or storage.

Project 2: Speech-to-Speech Converter (S2S)

We then combined two components — Speech-to-Text (STT) and Text-to-Speech (TTS) — to build a full Speech-to-Speech Converter.

How It Works - Speech-to-Speech Architecture

- 1. Speech-to-Text (STT) Using Whisper
 - User speaks into a microphone.
 - The speech is captured and saved as an audio file (usually .wav).

 Whisper, an automatic speech recognition model by OpenAI, transcribes the voice into clean and punctuated text.

2. Text-to-Speech (TTS) - Using xtts

- o The transcribed text is passed as input to the **xtts model**.
- o xtts generates a **new audio file** from the text in a synthetic voice.
- o The final audio is played and saved.

Republic Summary

Speech Input → Whisper (STT) → Text → xtts (TTS) → Speech Output

This full-cycle conversion mimics natural conversation and can be used in real-world systems like:

- Multilingual AI voice translators
- Assistive communication devices
- Interactive AI storytelling
- Custom voicebots and dubbing tools

Learning Outcome

By completing this integrated system, we learned:

- How to chain multiple AI models to build a voice-based application.
- How to manage **audio input/output**, transcriptions, and synthesized responses.
- Real-world use cases for voice-to-voice automation.