

Multi Speaker Transcription



Group2 KJYIHUAYYDS

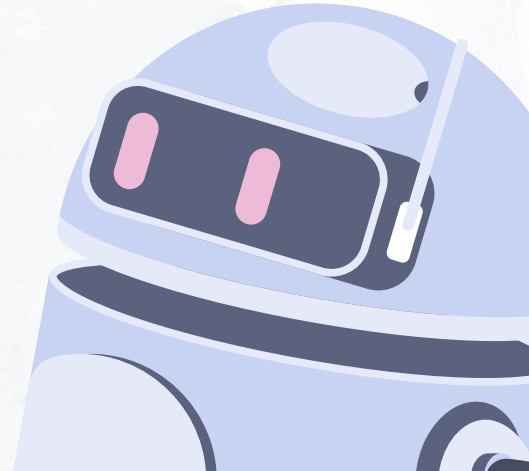


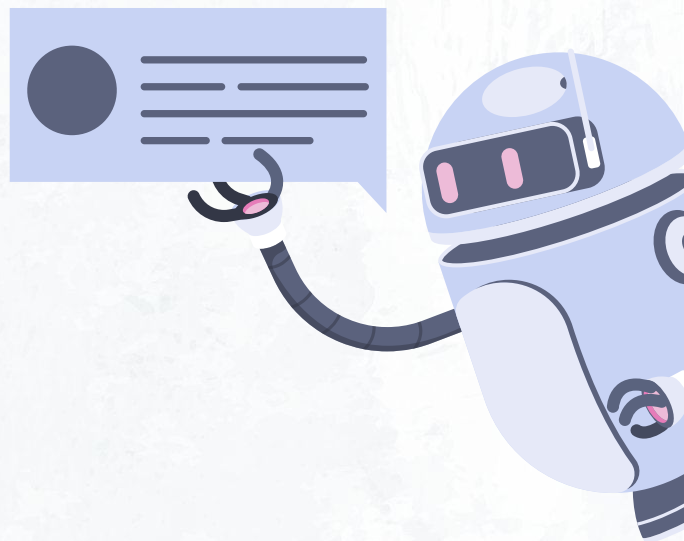
Table of contents

- 01 → Purpose
- 02 → Our solution-Back end
- 03 → Our solution-Front end
- 04 → Translation
- 05 → Demo

01



Purpose



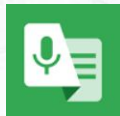
Purpose

Target Users →

- Low hearing people

Motivation →

- Current live transcript [application](#)



perform well on English

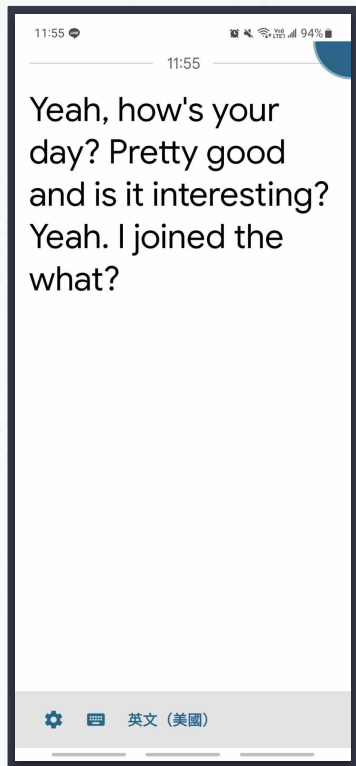
real-time transcript

What if it's a conversation with many people?

Setting



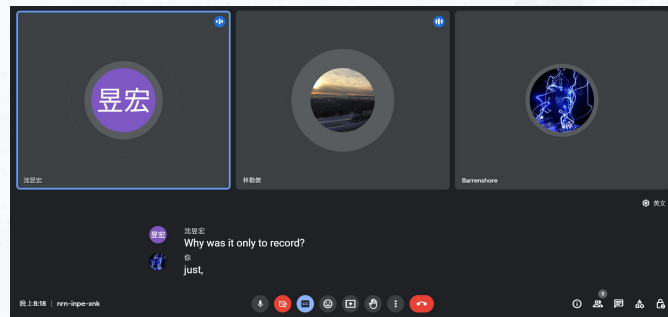
Example - 5-people conversation



Possible solution - voice identification

Hardware - google meet →

- not a good solution in our setting

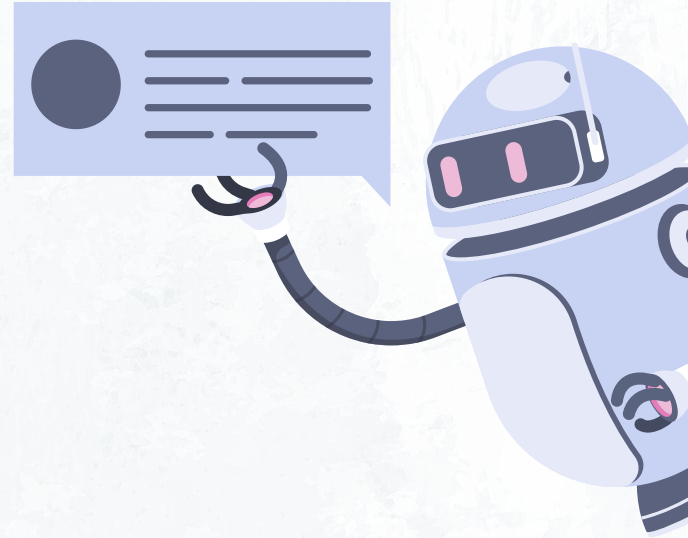


Software - speaker diarization(說話者分辨) →

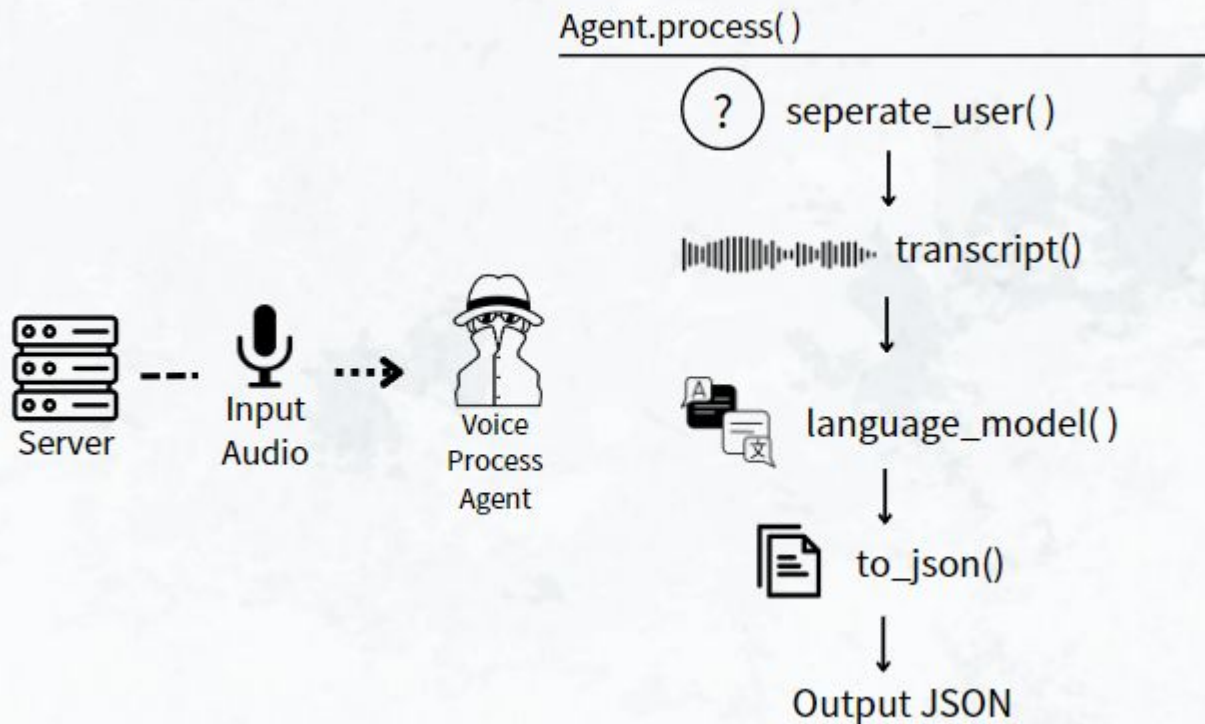
- Google cloud platform
- Amazon model
- often used for off-line data

02 →

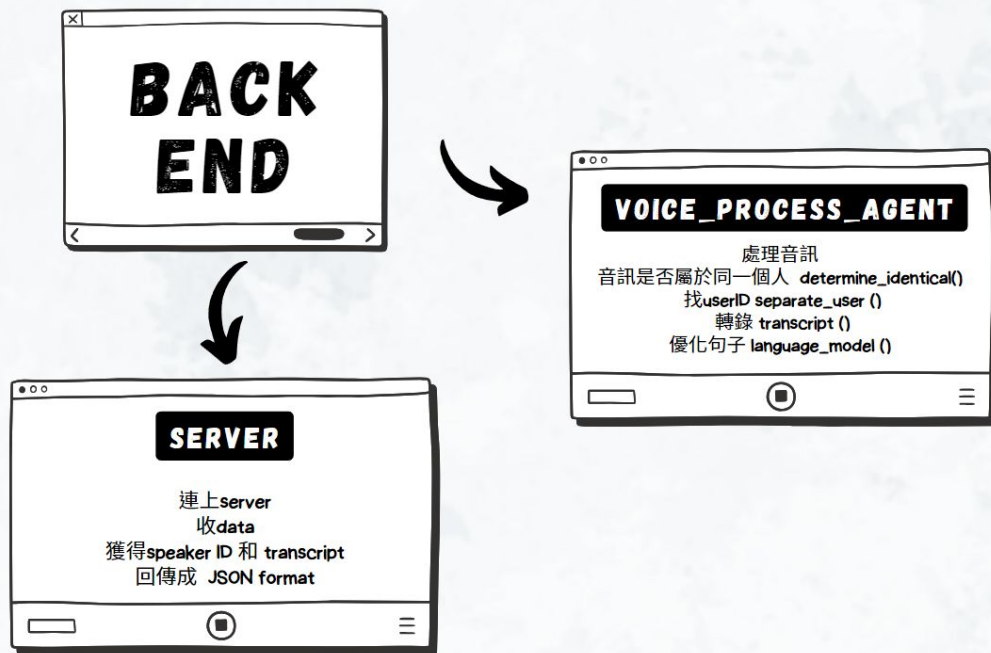
Our solution - Back end



Back end - concept



Back end - function



Models

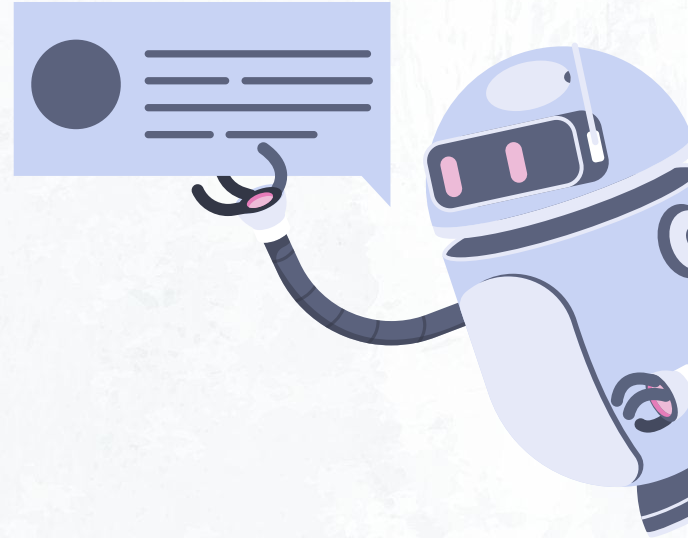
Separate user → speech brain

Transcript → speech_recognition library (in PYPI) (google)

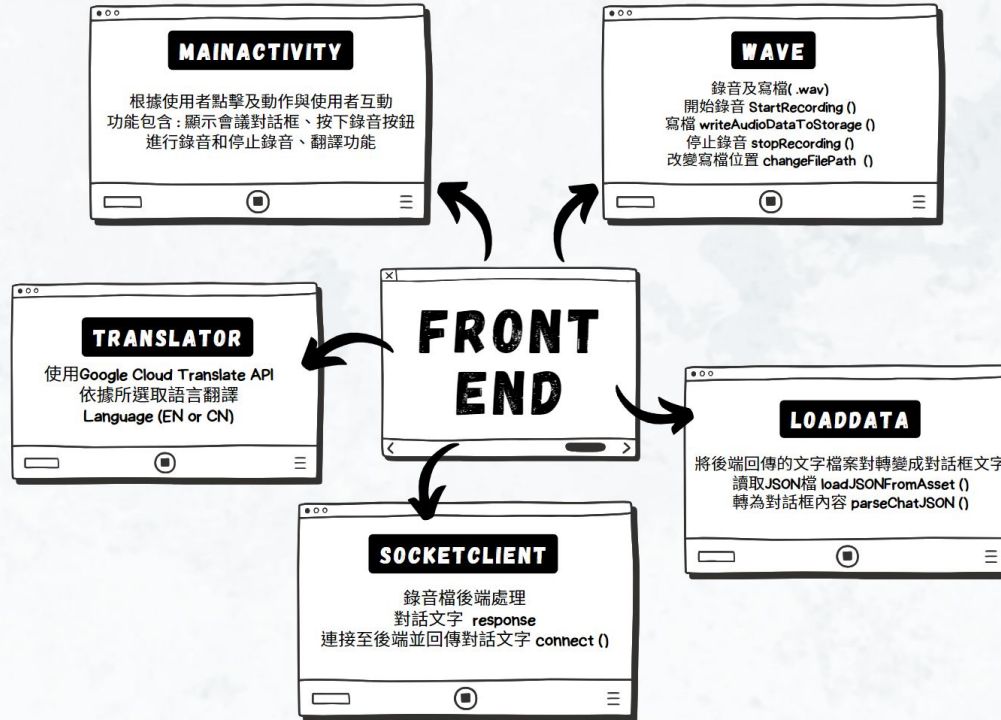
Language model → PALM

03 →

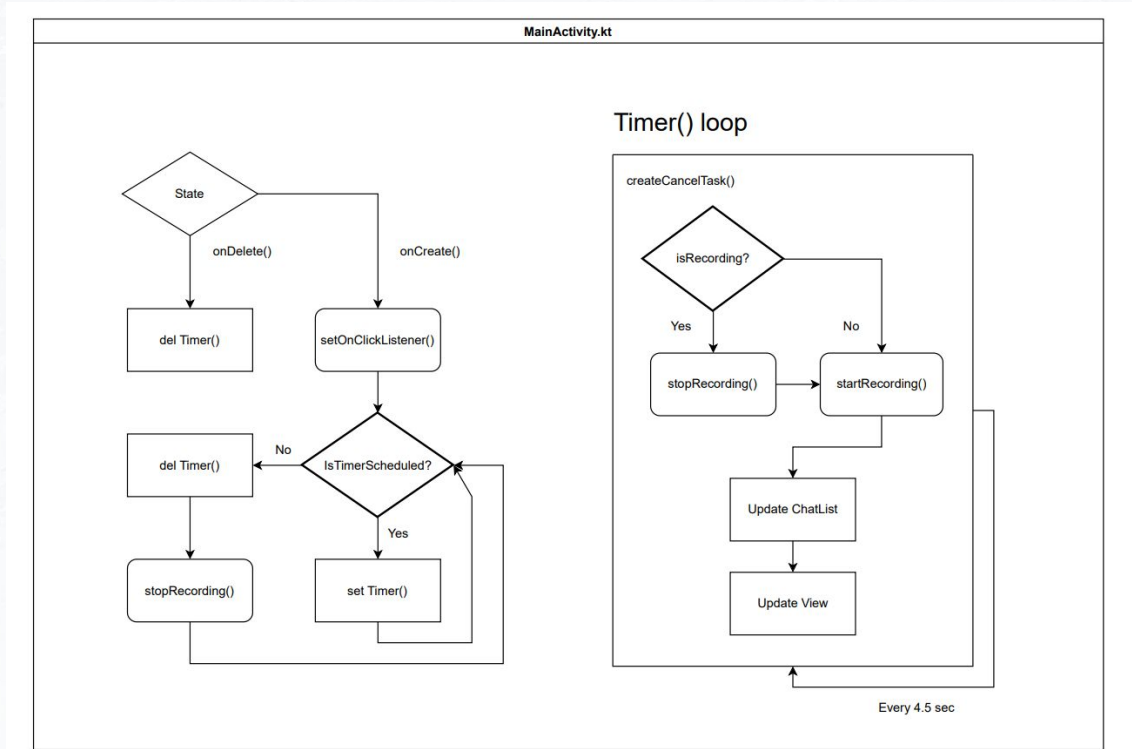
Our solution - Front End



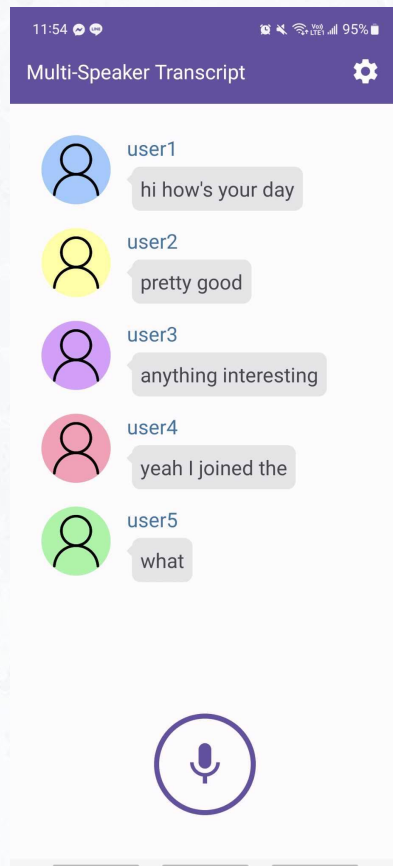
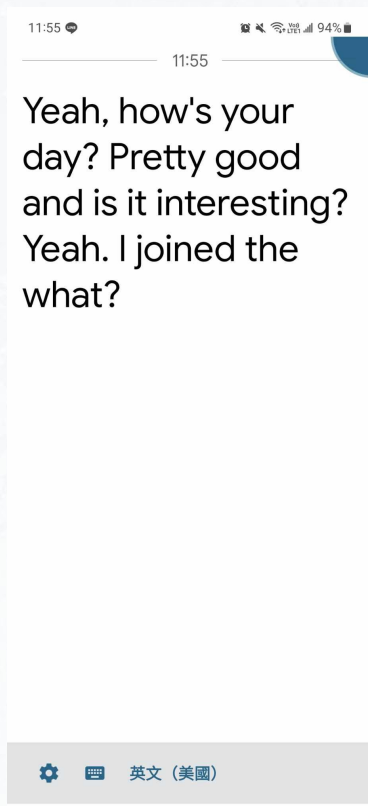
Front end - function



Front end - structure

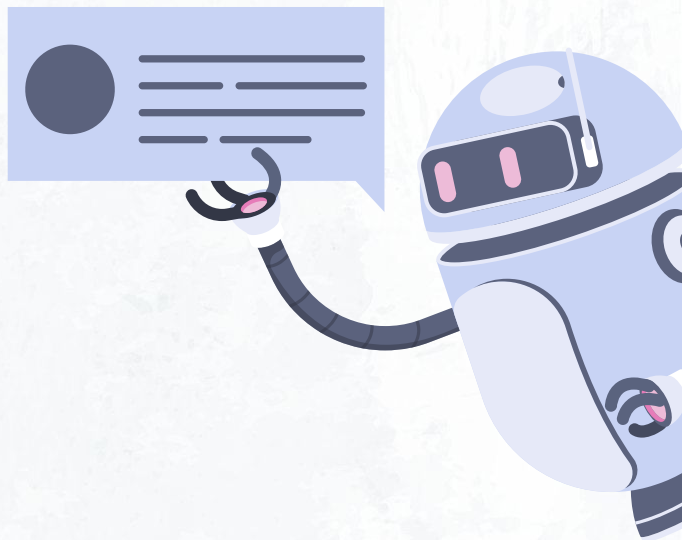


Result

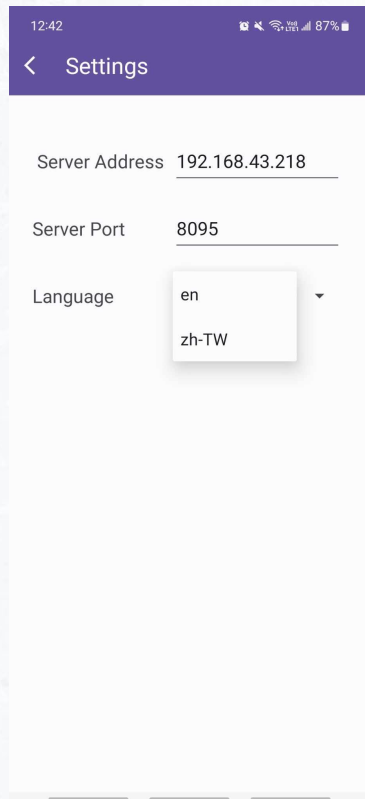


04 →

**Translation -
for more users**



Setting Layout

A screenshot of a mobile application's settings screen. The screen has a purple header bar with a back arrow and the word "Settings". Below the header, there are three settings: "Server Address" with the value "192.168.43.218", "Server Port" with the value "8095", and "Language" with a dropdown menu. The dropdown menu is open, showing two options: "en" and "zh-TW". The status bar at the top shows the time "12:42" and battery level "87%".

12:42 87%

< Settings

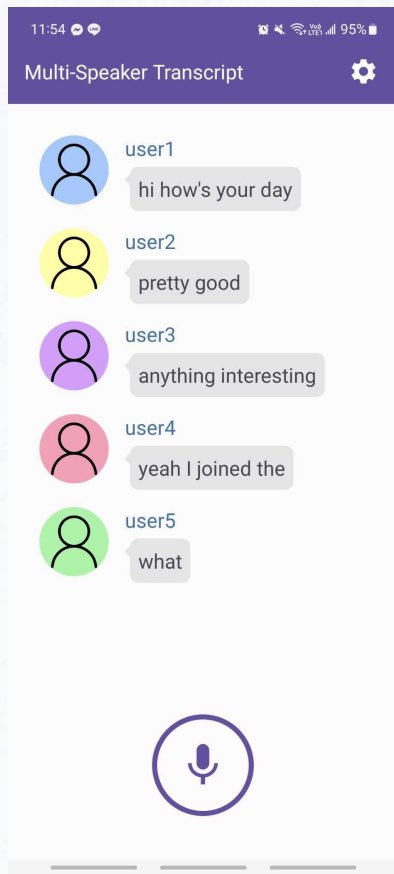
Server Address 192.168.43.218

Server Port 8095

Language

- en
- zh-TW

Result



05



Demo

