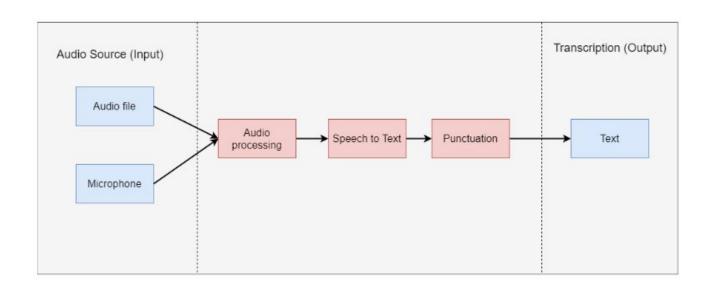
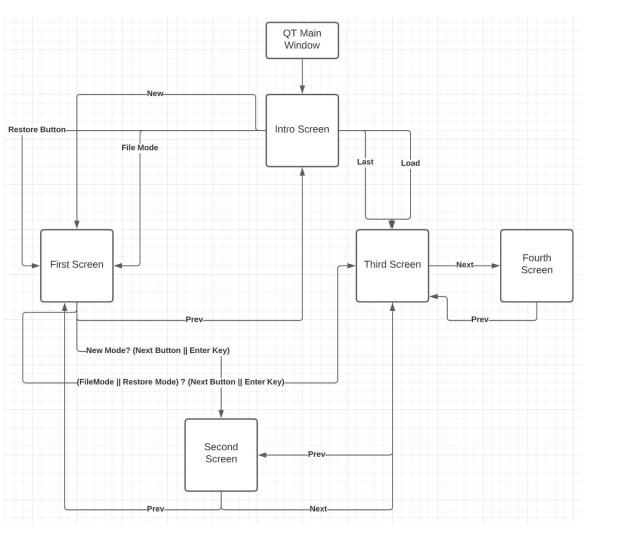
# Async ASR

# System Architecture

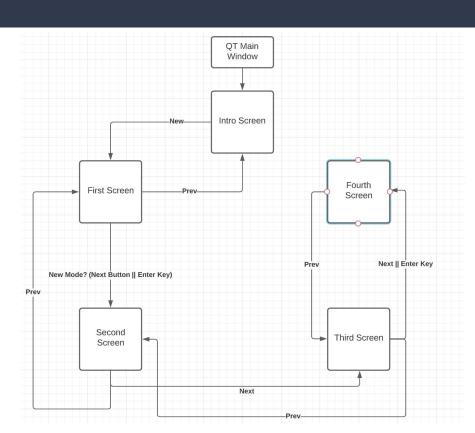


# Design Overview



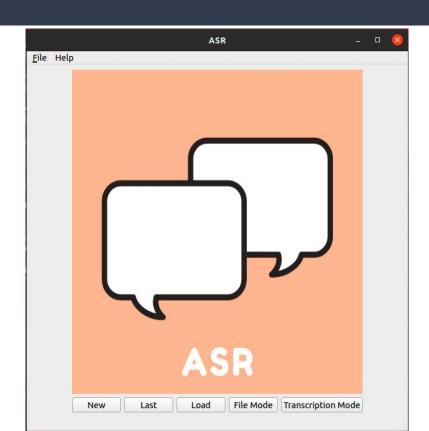
# New Mode

# Design



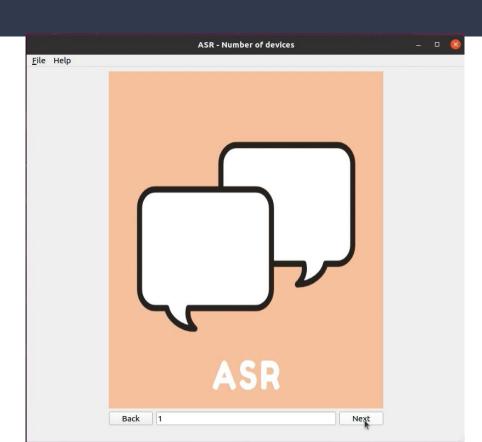
### New Mode

- Main Window is Declared
- Intro Screen UI is setup
- Intro Screen Attributes:
- ASR Logo
- New Button
- Last Button
- Load Button
- File Mode Button
- Transcription Mode Button



### First Screen

- > Selecting the new Button takes us to the first screen
- First Screen UI is set up
- Back Button takes us to the previous screen i.e. Intro Screen
- Enter the number of devices (max of 8)
- Next Button handles the number of devices and sets up the second screen by passing the number of speakers.

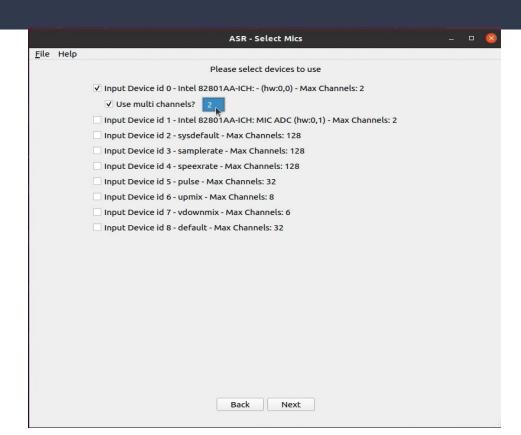


#### Second Screen

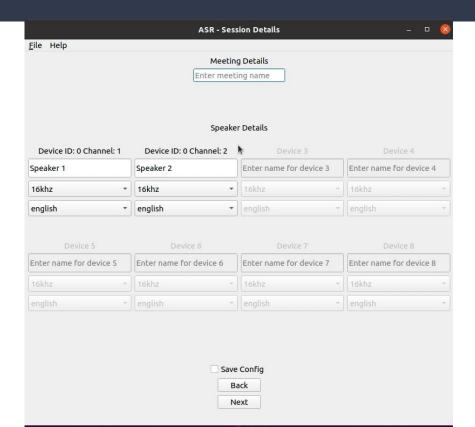
- > Clicking the new Button takes us to the second screen
- We add checkboxes for each of the device that we get from the function deviceInfo().

  When a checkbox is selected, a new check box is displayed if we want to use multi channels. if the checked, a combobox is displayed with the different channels that the device supports.

  Select the desired number of channels
- Next button sets up the UI for the third screen by passing the devices information to the third screen..



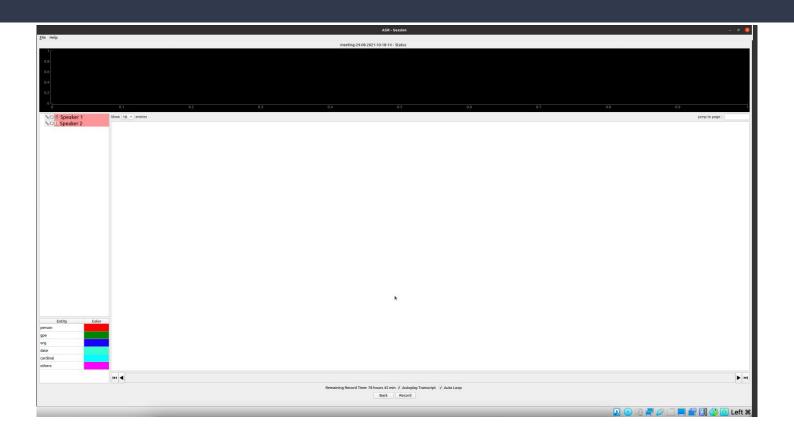
#### Third Screen



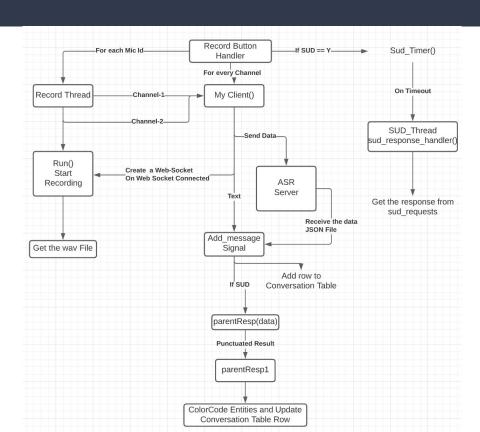
#### The third Screen UI is loaded

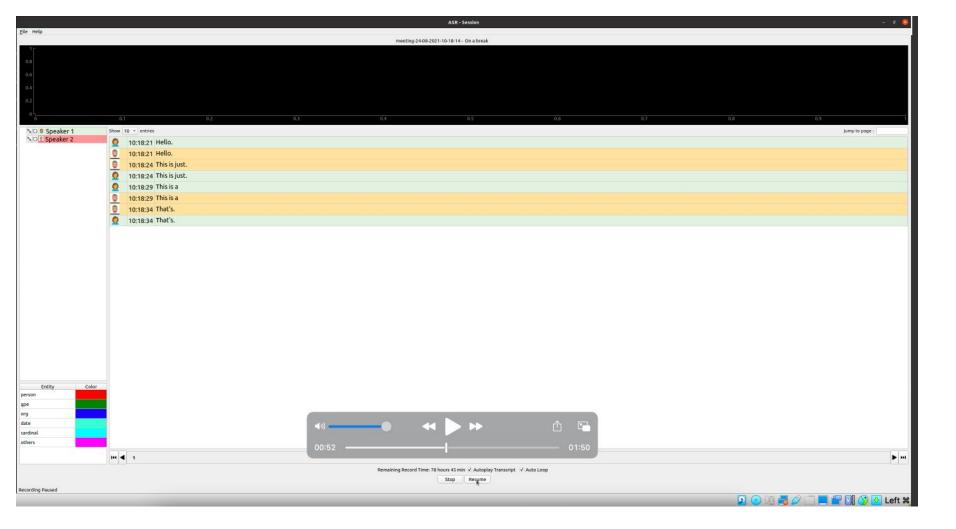
- -Enter the name of the meeting (Default: The current Date/Time)
- The Devices/Speakers are setup
- -The next button takes us to the fourth screen of the UI

## Fourth Screen

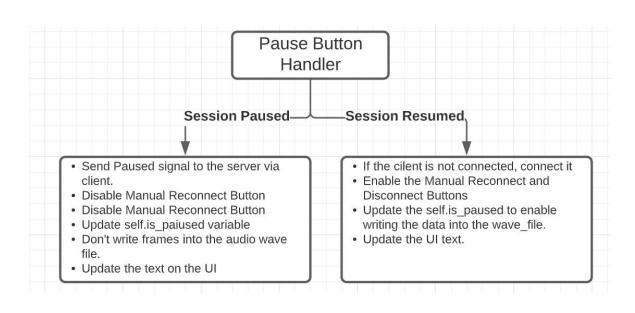


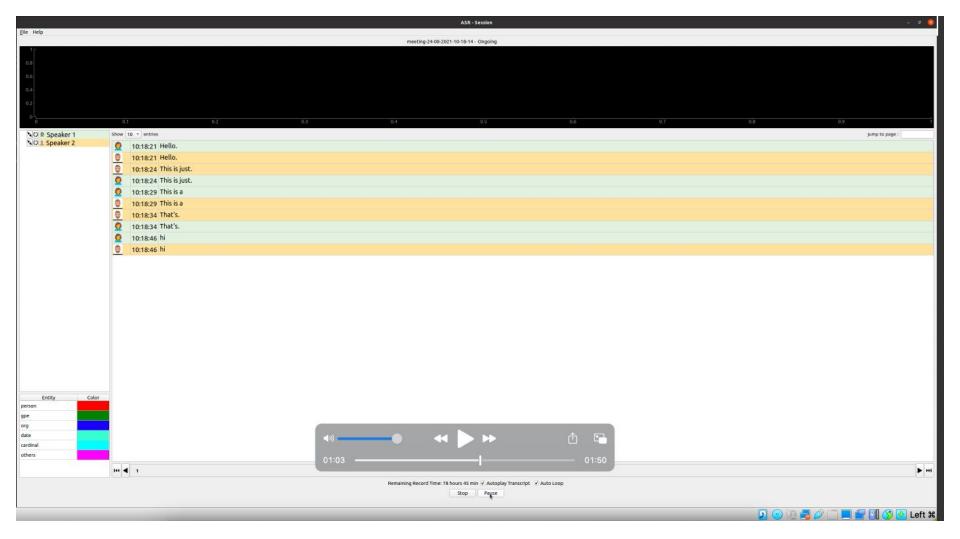
## Record Button Handler





### Pause Button Handler





# Stop Button Handler

Stop Button
Handler

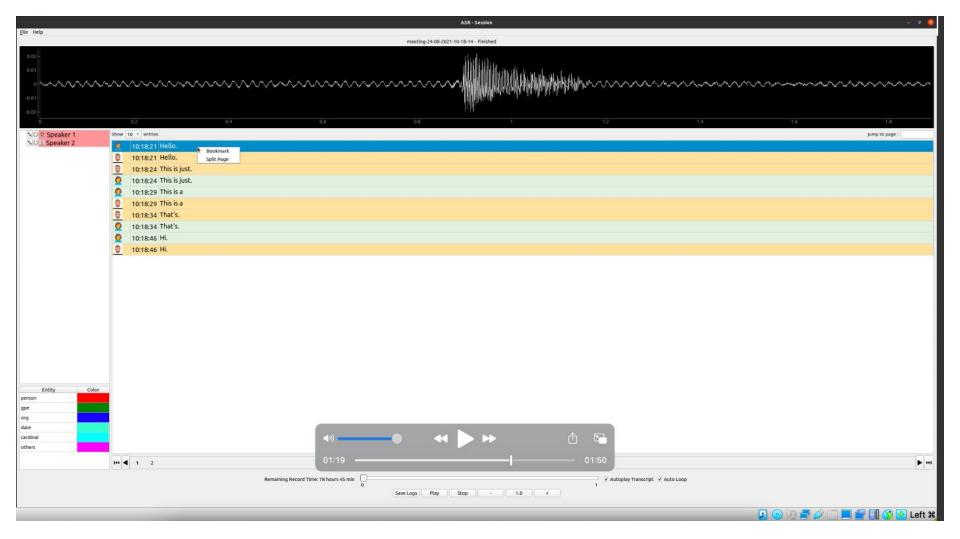
Stop the record\_time\_remaining\_timer
Stop the record\_threds for all the mic's
For every, mic thread, create a final way file.

- file

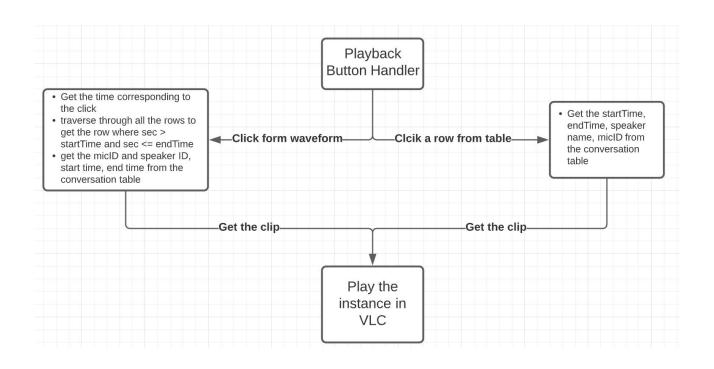
   Delete the part file
- If the file size exceeds the threshold set by us, save the file and create a new audio file for the next part files.

· Keep appending all the part files to the final

- Save all the file-names corresponsing to the mic-channel
- Update the UI by showing hte graphic view box.



# Playback Button Handler



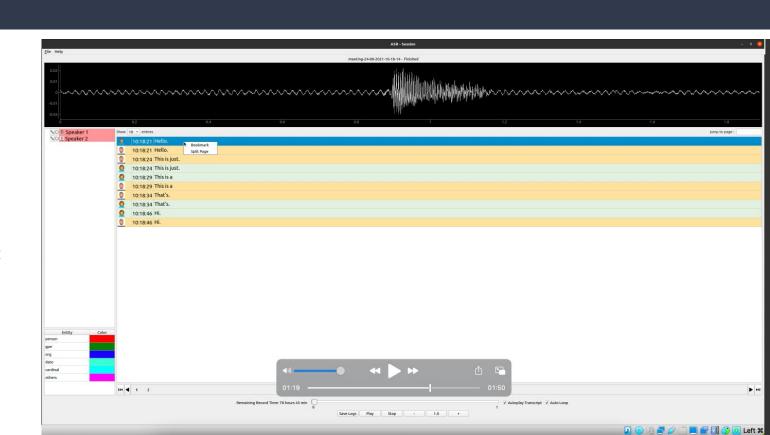
Increase\_playrate(): + Decrease Playrate(): -Reset Current Rate():



# Paging

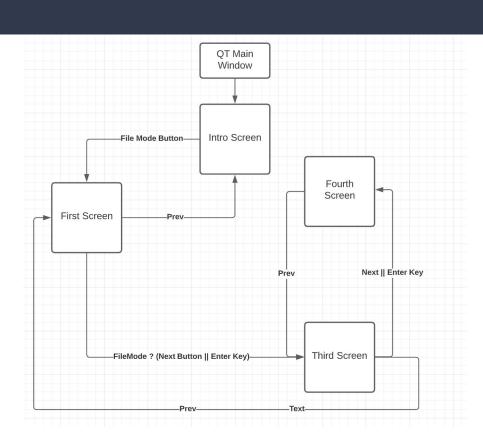
Prev Button Handler():
Skip First Button
Handler():
Skip Last Button():
List Item Clicked():
Update Page Limit():
Page Jump():
Right Click event filter():
BookMark():
Split Page():

Next Button Handler():



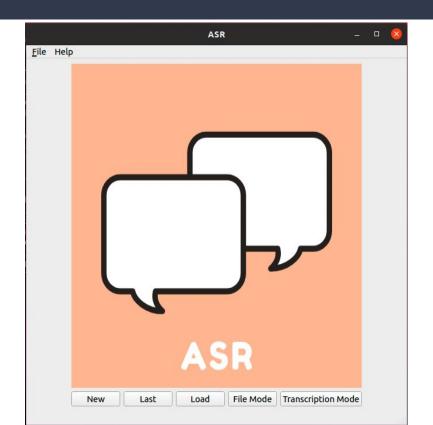
# File Mode

# Design

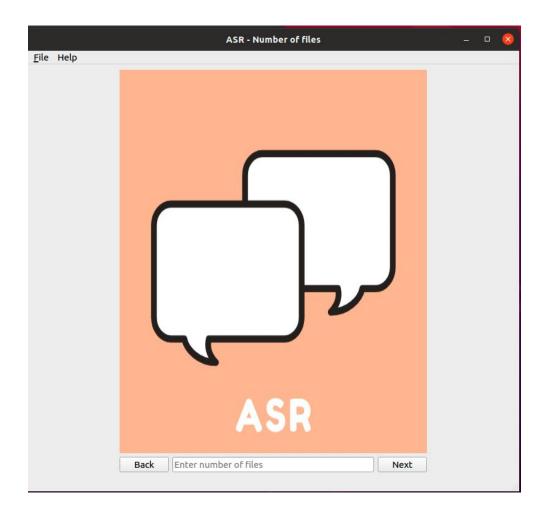


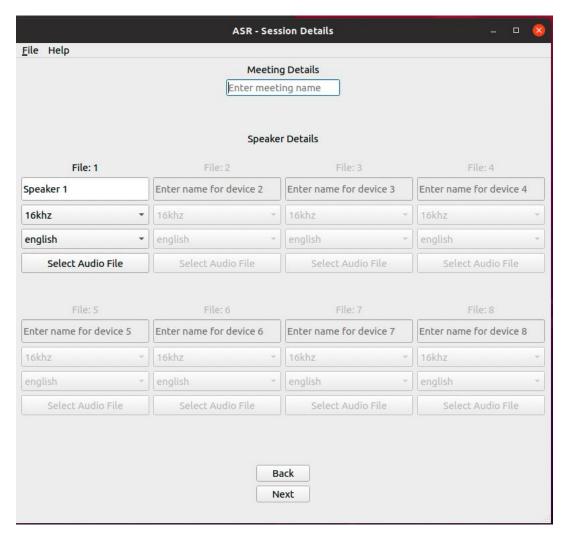
## File Mode

- Main Window is Declared
- Intro Screen UI is setup
- Intro Screen Attributes:
- ASR Logo
- New Button
- Last Button
- Load Button
- File Mode Button
- Transcription Mode Button



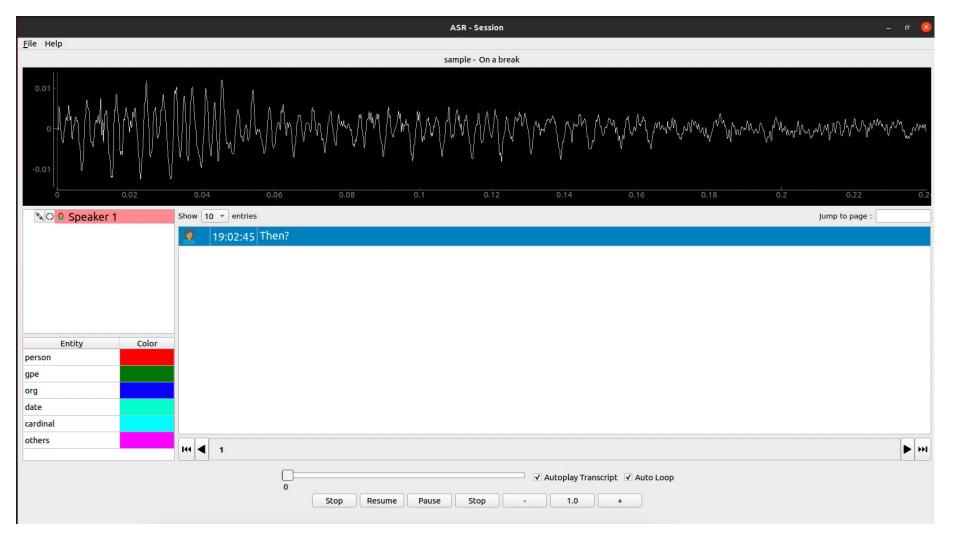
- > Selecting the new Button takes us to the the first screen
- First Screen UI is set up
- Back Button takes us to the previous screen i.e. Intro Screen
- Enter the number of files (Max 8 Files)
- Next Button handles the number of devices and sets up the second screen by passing the number of files.





#### The third Screen UI is loaded

- -Enter the name of the meeting (Default: The current Date/Time)
- The Devices/Speakers are setup
- The audiofile for that speaker needs to be selected.
- -The next button takes us to the fourth screen of the UI



# Async ASR

# Multi-Channel Audio File Support

- The system currently has a limitation whereby the user is unable to transcribe audio files with multiple channels. The user is required to split the channels into different audio files manually to process the audio file.
- The application can automatically split the audio file with multiple channels into separate audio files and process them.

#### Library: wave, numpy or spicy.io

- Get the number of channels in the audio file (wav.getnchannels())
- For each channel, store the channel data into a different file with the name given by the user/ speaker name.
- Use the required channel audio files as an input.

# Conversation Table Resize

#### Problems:

- The Row number of the table is used in the payload of the ASR and the SUD services.
- We pass the row number and the data of the row is updated in on receiving the hypothesis.
- Updating the row number will cause errors in the above services.

#### Alternative:

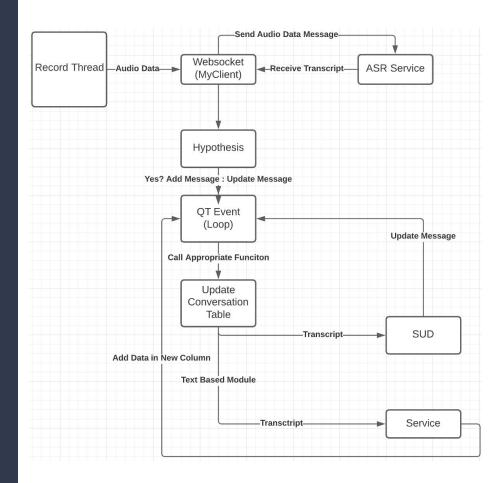
- Add a horizontal scroll bar. (Will need to lookup about this).

# Using a Speech Services

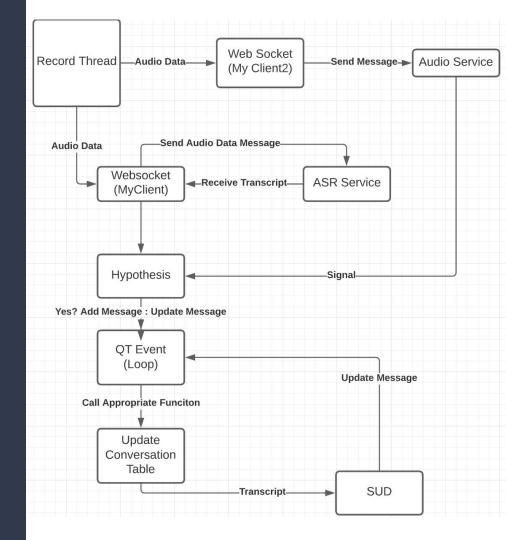
#### 1. HTTP requests

- Requests Library
- Futures Add-on to requests Library.
   (Create an instance of Futures Session)
   Helps in asynchronous communication.
- Whenever we make a http request, we add the request instance to the queue.
- Every 10 seconds, we check if the request is fulfilled. Else, we resend the request.
- 2. Server-Client based communication.
  - Create a web-socket for the communication.
  - Send the message through the web-socket
  - Create a signal to process the message when received

# Text Based Speech Module

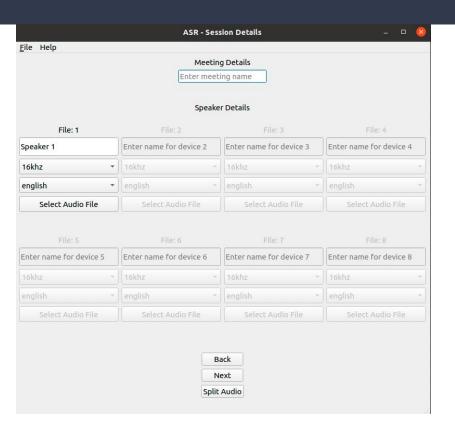


# Audio Based Speech Module



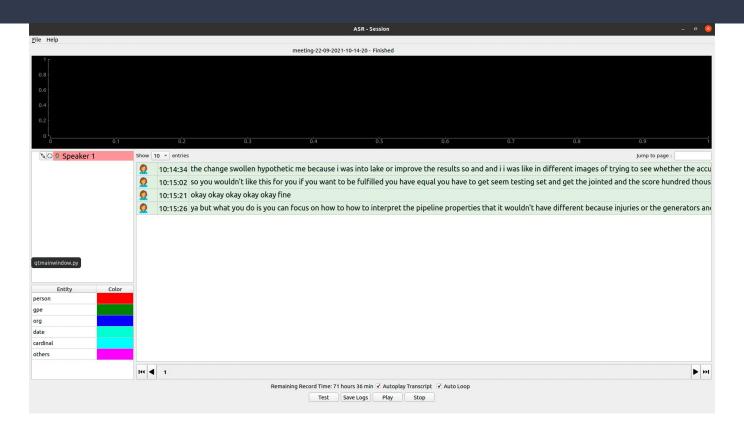
# Asynchronous Automatic Speech Recognition

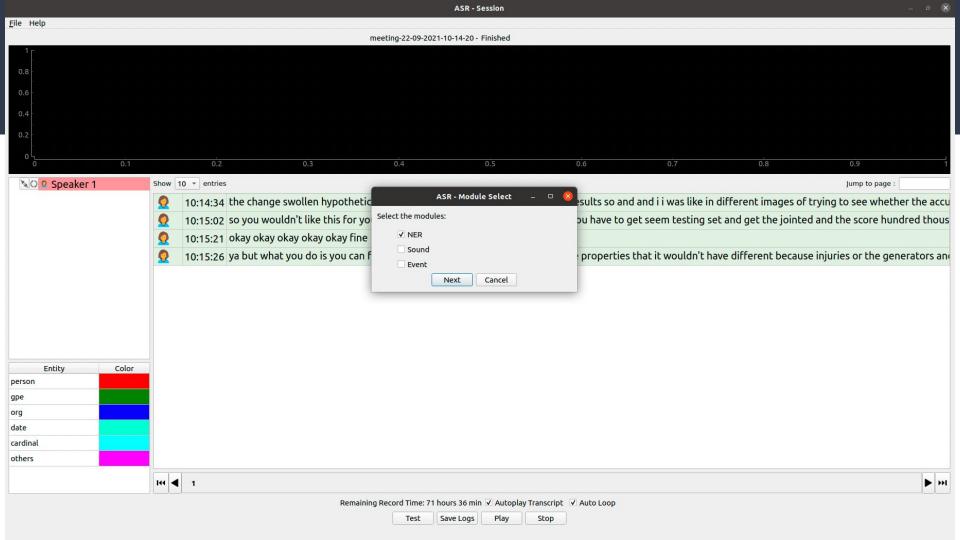
# Split Audio with Multiple Channels.



- Split Audio Button Opens a File Selector Window
- Select the Audio with multiple channels to split into individual audio files.
- SOX module splits the audio file into multiple audio files, one file for a channel
- If the audio file is file.wav, the split audio files are saved as file-channel-1.wav, file-channel-2.wav ... etc
- The audio files are stored in the temp folder in the current directory.

# ASR - Module Analysis







## Current Problem

I am not able to use the Analyze button more than once while running the project.

Error:

RuntimeError: wrapped C/C++ object of type QTableWidgetItem has been deleted