

Smart Voice Assistant for Library System

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ABSTRACT: Library plays an important role in student's life. It is also helpful for the research scholars. Today libraries are established in all places. Finding a book in the library is a herculean task. Some people even loose interest while searching the required book. Searching for books in computer also takes much time. In our project we use voice recognition to find books which will make the herculean task easier. Even the people without system knowledge can access this and find books in an easier way. It reduces the time in searching books and makes the library user friendly.

KEYWORDS: iot , raspberry pi 3 model b , Speech recognition , speech to text process

I.INTRODUCTION

Library is a huge collection of books. The proper arrangement of books in given collection of order that makes it simple for the user to locate a specific book. This process used in large libraries having a huge collection. They are not efficient and managed manually. The time has been wasted by searching for the book in the library. This problem mostly occurs in libraries, where number books is enormous. Thus, the fact of all books are arranged in classifications, the area of classification should first be known to the user becomes more acquainted with certain specific class of books is placed. In such a condition there should be a need through client can definitely find the area of any book by simply saying its name.

In this proposed system we use a microphone and speaker to establish interactive communication between the user and the system. With the speech recognition library, we can process the user's query and respond by providing the requested information to the user. Speech recognition is the ability to receive and interpret dictation to understand and carry out given commands. Voice recognition has gained prominence and use with the rise of AI and intelligent assistant. Voice recognition systems enable user to interact with simply way of speaking.

This technology provides smart searching for finding books from the libraries in an efficient way rather than finding or searching the books manually as in the conventional or traditional methods. Searching books manually which will take more time to find a book. This device will help to search and find the book quickly and efficiently. As this device is using voice recognition technology it helps in searching books very easier than the manual searching.

A. EXISTING SYSTEM

In the libraries the books are searched and founded manually. People will search the books which they wanted specifically from catalogue which is produced by the library. After finding the book from the catalogue the need to note the book's location number or any other locating techniques of the libraries. So, by searching the locating number they could able to find the book. If the people can't able to find their book in the catalogue, they have to ask the librarian for getting the book.

B. LITERATURE SURVEY

This paper addresses different configurations of two layer and three layers neural networks approach for the low resources like gujarati. ear microphone recorded up to 2.3kHz-4kHz Noise hum -90dB around 200 Hz.. MFCC and Real Cepstral Coefficient (RC) Conjugate Gradient (CG) Algorithm and Levenberg-Merquardt (LM) Algorithm are used. This can be accomplished by discourse acknowledgment framework, which allows an information processor to distinguish the words an individual expresses in a receiver or phone, and changes over them into written text Television broadcasts containing both male and female speaker recorded and 10 Tamil spoken digits (0-9) and 5 spoken names from 30 different speakers Sampling rate 16 kHz .

II..PROPOSED SYSTEM

This system works on speech recognition technology. It uses raspberry pi instead of using costly computers. This process will convey the message through USB speaker. It is used as an individual device for libraries. This process will replace the man work to system work and also reduce the time consumption. By using this concept user can find the books in easy way without any trouble and tension it won't reduces the User interest . The location of the book will be guided through Speaker . By this way the user will find the book location easily .

A.Materials and Methods

USB speakers & microphones: It doesn't require audio drivers used by other speaker systems; rather, they use the system's USB drivers. In effect, USB speakers and mic include a built-in sound card. This makes them an excellent replacement option for a malfunctioning or inoperative sound card.

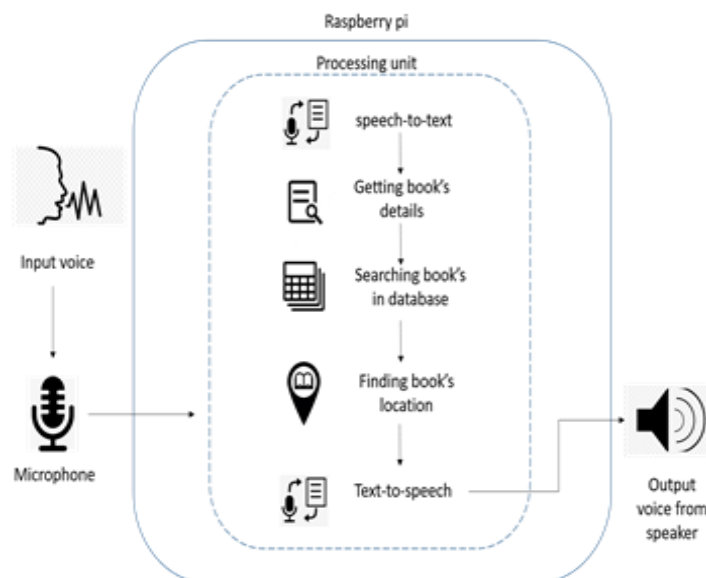


Fig.1 Proposed system architecture



Fig.2 Microphone



Fig. 3 Speaker

2. Speech recognition: The first component of speech recognition is, of course, speech. Discourse must be changed over from physical sound to an electrical sign with a mouthpiece, and afterward to advanced information with a simple to-computerized converter. Once digitized, a few models can be utilized to interpret the sound to content. To disentangle the discourse into content, gatherings of vectors are coordinated to at least one phonemes—a central unit of discourse. This figuring requires preparing, since the sound of a phoneme differs from speaker to speaker, and even fluctuates starting with one expression then onto the next by a similar speaker. A unique calculation is then applied to decide the most probable word (or words) that produce the given grouping of phonemes.

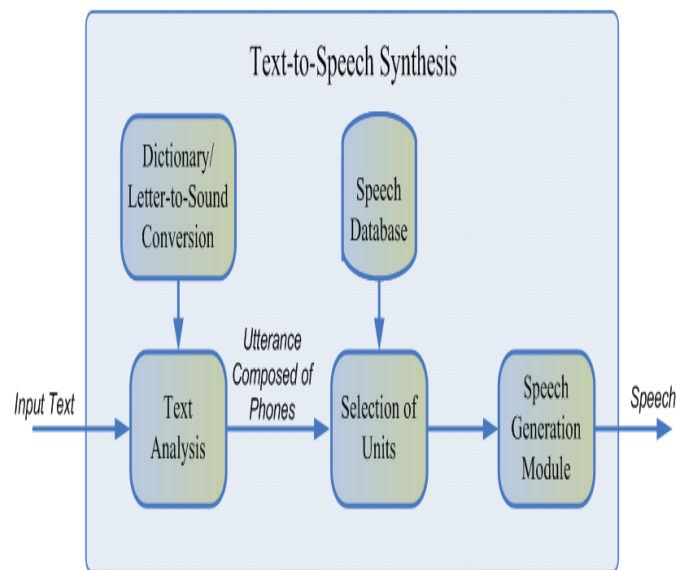


Fig.4 Text to speech

3 .Raspberry Pi 3: This board is a miniature marvel, packing considerable computing power into a footprint no larger than a credit card. It's prepared to do some astonishing things, yet there are a couple of things you're going to need to know before you dive straight into the thorn fix.

Since you have a fundamental comprehension of how the Pi varies from other figuring gadgets, it's an ideal opportunity to begin.. If you've just received your Pi, take it out of its protective anti-static bag and place it on a flat, nonconductive surface before continuing with this chapter. This was proposed to be exceptionally reliant upon task stringing and guidance set use. Benchmarks showed the Raspberry Pi 3 to be approximately 80% faster than the Raspberry Pi 2 in parallelized tasks. The Raspberry Pi Foundation uses Raspbian OS.

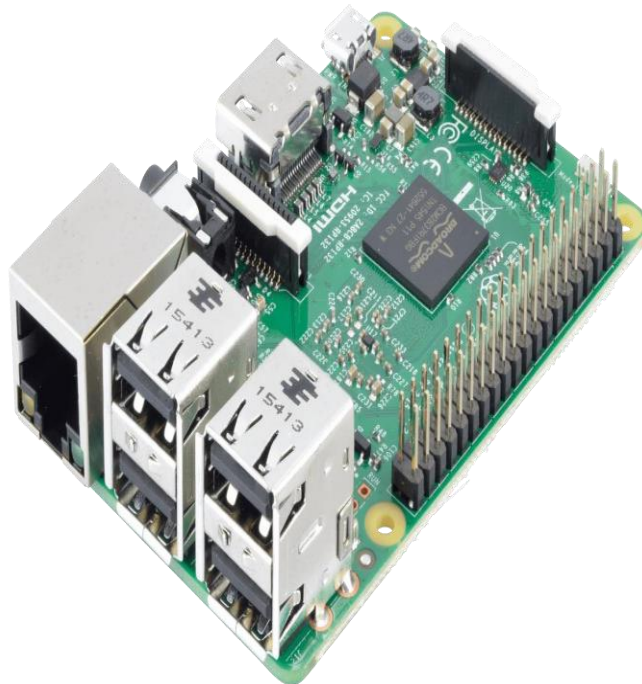


Fig.5 Raspberry Pi 3(MODEL B)

3.SYSTEM DESIGN

This system design consists of three units 1. Microphone (input unit) 2. Raspberry pi 3 (processing unit)
3. Speaker (output unit)

A. INPUT UNIT

The person will convey the command in voice through microphone .It is easy to get identify the author of book name and section.

B. PROCESSING UNIT

In this processing unit is the raspberry pi. The raspberry pi are gets the input from the USB microphone. Speech must be converted from physical sound to an electrical signal with a microphone, and then to digital data with an analog-to-digital converter and it convey the output through the USB speaker.

C. OUTPUT UNIT

After the processing unit, this unit will convey the command to speaker and its easily recognize by that person.

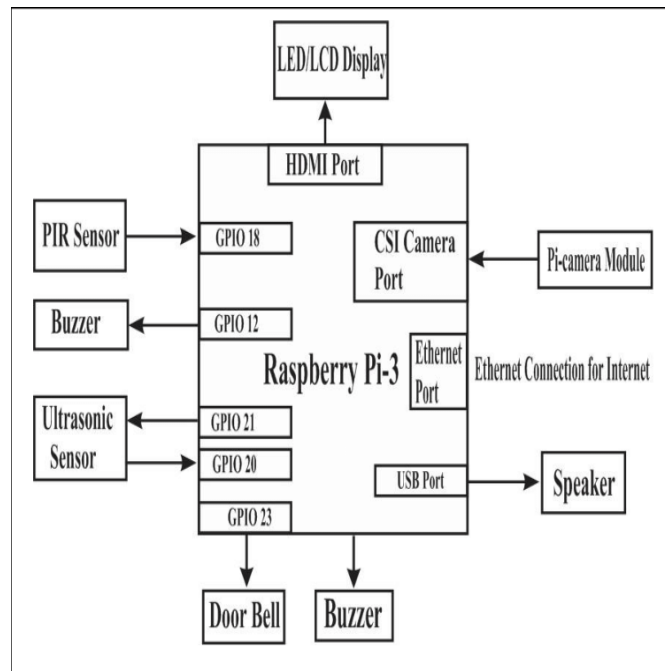


Fig.5 Block diagram

IV.EXPERIMENT AND RESULT

Thus , the raspberry pi help to interact with human for correct way of Speech recognition . The ability of a machine learning to identify words and phrases in spoken language and convert them to a machine-readable format. More sophisticated software has the ability to accept natural speech this way convey the location command to the user to identify the book.



Fig.6 Output

V. CONCLUSION

Thus, in this paper, the objective of the proposed system has been achieved. The vision of this project is to search books in libraries without any difficulties instead of searching the manually. Smart library management system is developed using a voice recognition technology which is very helpful for searching the books very effectively and it saves more time and reduces the difficulties for searching and finding the books. As the system is using raspberry pi it is very efficient and very convenient for accessing and working. Hence a new technology has been proposed it will be very for helpful for future upcoming generate.

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