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A PROPOSAL ON LIP READING USING CONVOLUTIONAL NEURAL NETWORKS

Submitted in partial fulfillment of the requirements for the degree

BACHELOR OF COMPUTER ENGINEERING

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Abstract

Lip reading is the task of decoding text by interpreting the movements, shape and other spatiotemporal facial features from a recorded video clip of a speaker. It invloves mainly studying the movements and configurations in and around the lip area of a speaker. It can be especially useful for people with speech and hearing disabilities so that they can better convey their message to a listener. Besides that, it can be used as a means of comprehending or captioning spoken media from videos recorded in situations where sound may be difficult to perceive, such as in noisy environments or when the speaker is at a distance. In recent years, technological advancements, particularly in the field of computer vision and machine learning, have led to the development of automated lip reading systems. These systems use algorithms and models to analyze lip movements and convert them into text, providing potential applications in areas such as assistive technologies, human-computer interaction, and surveillance. This proposal outlines a comprehensive research project aimed at advancing the field of lip reading through the integration of Convolutional Neural Networks (CNN) and Long Short-Term Memory (LSTM) networks. The proposed hybrid architecture leverages the strengths of both CNNs and LSTMs to enhance the accuracy and efficiency of lip reading systems, addressing the inherent challenges in visual speech recognition. The proposed hybrid architecture will be trained on a comprehensive dataset, including diverse speakers, languages, and environmental conditions, to ensure robustness and generalization. Fine-tuning mechanisms will be implemented to optimize model parameters and improve its adaptability to various lip reading scenarios.

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List of Symbols and Abbreviation

CNN Convolutional Neural Networks ASR Automatic Speech Recognition

ALR Automatic Lipreading

Bi-GRU Bi directional Gated recurrent unit

RNN Recurrent Neural Network SAT Speaker Adaptive Training

Introduction

1.1 Background

People often communicate through hearing and vision, that is, through voice signals and visual signals. Speech signals often contain more information than visual signals, so many studies have focused on Automatic Speech Recognition (ASR). Although automatic speech recognition (ASR) technology is mature, there are still some unsolved problems, such as how to accurately identify what the speaker is saying in a noisy environment. Lipreading is a visual speech recognition technology that recognizes the speech content based on the motion characteristics of the speaker's lips without speech signals. Therefore, lipreading can detect the speaker's content in a noisy environment, even without a voice signal. Machine learning methods have a great impact on social progress in recent years, which promoted the rapid development of artificial intelligence technology and solved many practical problems. Automatic lip-reading technology is one of the important components of human-computer interaction technology and virtual reality (VR) technology. It plays a vital role in human language communication and visual perception. This project investigates the task of speech recognition from video without audio. The input data to our algorithm is sequences of still images taken from frames of video. We use models to output one of 10 words that are spoken by a face in the input images. We explore and combine a number of different models including CNNs, RNNs and existing publicly available pretrained models to assist in mouth recognition.

1.2 Problem Statement

At present, the ASR can reach a very high recognition rate without severely damaging the speech signal and also can be used in many practical fields. Visual speech recognition is a technology that recognizes the speech content by lip movement characteristics on no speech signal. The information received by the voice channel is two dimensional. Compared with the one-dimensional voice information received by the voice channel, the visual information often contains more redundant information. So visual speech recognition has always been a difficult problem to solve. Visual speech technology is also known as Automatic Lipreading (ALR), which infers the speech content according to the movement of lips in the process of speaking. In real world, there are people with hearing impairment. They communicate through sign language or observing through people's lip movements. But gesture language has problems such as being difficult to learn and understand, and inadequate expression skills. Therefore, ALR technology can help people with hearing impairment communicate with others better to some extent. Also in noisy environments, the speech signal is easily interfered with by the surrounding noise, resulting in the reduction of recognition rate. However, the visual information needed for ALR will not be affected, so ALR can improve the recognition effect of speech recognition in noisy environments. In the field of security, first of all, with the popularity of face recognition technology, there are many attacks against face recognition system, such as photos, video playback, and 3D modeling, etc. adding lip features can further improve the security and stability of the security system. In the field of vision synthesis, traditional speech synthesis can only synthesize a single voice, and lipreading technology can generate high-resolution speech scene video of specific people. Besides, in sign language recognition, lip movements are also combined to better understand the content of sign language or improve the accuracy of sign language recognition.

1.3 Objectives

The main aim of this project is:

- To help hearing impaired people.
- To improve the accuracy and stability audiovisual applications.
- To extract proper data from any of the noisy environment.

Literature Review

The introduction of Artificial Intelligence has greatly enhanced the interaction capabilities of people with hearing and speech related disabilities and impairments. With there being millions of people suffering from these disabilities, the use of suitable lip reading applications and models can allow them to engage in conversations, thus making them be connected to the real world. However, developing such a model is challenging for both designers and researchers. These models should be well designed, perfected, and integrated into smart devices to be widely available to all people in need of speech understanding assistance.

Lip reading can be conducted on the letter, word, sentence, digit or phrase level. It can also be based on video, voice, video with voice or video without voice as input. There have been studies focused on speaker-independent lip reading by adapting a system using the Speaker Adaptive Training (SAT), which was initially used in the speech recognition field. [2]. Research has also been done towards developing an audio-visual speech enhancement framework that operates at two levels: a novel deep-learning based lip-reading regression model and an enhanced, visually-derived Wiener filter for estimating the clean audio power spectrum. [1] The paper [5] uses CNN and Bi-GRU(Bi directional Gated recurrent unit). According to this algorithm, the system is decomposed into two blocks. The first block consist of lip segmentation. The mouth region is extracted using Haar Cascade classifier. Then hybrid active contours model with an improved of the edge by a designed filter is proposed. The second block consists to classify word lipreading. First, deep convolutional neural network (CNN) is applied to extract frame features from videos who take the results of first block as inputs. Second, the Bi-GRU with two hidden layers is followed by a global average pooling layer. Finally, the word classification results are obtained by Softmax layer. Using segmented lip inputs can yield stronger features, and vastly improve recognition performance. The paper [3] proposes a novel lip-reading driven deep learning approach for speech enhancement that leverages the strengths of deep learning and analytical acoustic modeling. The proposed audio-visual speech enhancement framework operates at two levels: a novel deep learning based lip-reading regression model and an enhanced, visually-derived Wiener filter for estimating the clean audio power spectrum. This discusses the challenges of lipreading and presents LipNet, a model that can map a sequence of video frames to text, trained entirely end-to-end. On the GRID corpus dataset, LipNet achieves 95.2% accuracy in sentence-level, overlapped speaker split tasks.

The [4] uses the MIRAVL-VC1 dataset which outperforms previous datasets in various aspects. It uses modified form of residual network architecture and uses various techniques in data processing, augmentation and visualization to overcome the scarcity of data and improve the performance. Possible insight into possible improvements and future work in expanding the scale and generalization of the model.

Feasibility Study

3.1 Technical Feasibility

The technical feasibility of a lip-reading application falls on usage of advanced image and video processing techniques in order to capture and process and analyze clear lip movement. The system should be seamlessly running with speech-to-text capability. And usage of natural language processing is a must in order to increase the transcription ability. Real-time processing capability and consideration of hardware requirements is a must for bringing the system into practical use. Through the use of large and varied dataset using an effective model and also keeping the functioning of system in different lighting conditions in mind a consistent user experience is expected.

3.2 Economic Feasibility

The economic feasibility of a project verifies project's financial viability by examining a project's costs, benefits, and risks to determine whether it is financially viable and worthwhile to pursue. For this system, the economic feasibility would involve the cost of training and fine-tuning multiple image and video processing models, implementation of the project its software development keeping required hardware in mind along with its maintenance cost.

3.3 Schedule Feasibility

The scheduling feasibility of a project is an assessment of whether the project and be completed within a specified time frame maintaining quality standard. There are several factors which could impact the schedule, including the availability of the resource materials the project is estimated to take a little over then 3 months. There is expectations to finish the documentation and testing of the system in the specified time frame.

Project Methodology

4.1 Software Development Model

The Agile model is an adaptable and iterative software development process that puts the needs of the client and flexibility first. It breaks the project up into manageable chunks known as sprints or iterations, enabling regular review and modification. Close collaboration between cross-functional teams results in functional software at the conclusion of each iteration. This cycle of iteration guarantees prompt reaction to evolving needs, promoting ongoing enhancement and contentment for the client. The Agile Manifesto's concepts of agile development include a strong emphasis on people and their relationships, functional software, customer collaboration, and adapting to change. In dynamic development contexts, the Agile approach has gained widespread adoption as a framework that encourages efficiency and reactivity.



Figure 4.1: Agile Model

4.2 Block Diagram of proposed system

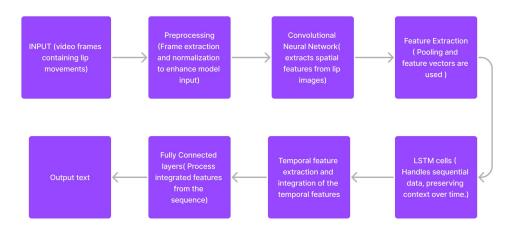


Figure 4.2: Block Diagram of Proposed System

4.3 Description of working flow of proposed system

1. Input(Video Frames):

The system takes video frames containing lip movements as input, forming the basis for lip reading analysis.

2. Preprocessing:

Frames undergo extraction and normalization to enhance the model's input quality, ensuring consistent and standardized input data.

3. Convolutional Neural Network (CNN):

A CNN is employed to extract spatial features from lip images by utilizing convolutional layers with filters. These filters detect spatial patterns within the lip movements.

4. Feature Extraction:

Pooling layers are used to reduce spatial dimensions, and the result is feature vectors that effectively represent distinctive lip features.

5. Recurrent Neural Network (RNN):

RNN processes temporal sequences of features, allowing the model to capture temporal dependencies within the lip movements.

6. Long Short-Term Memory (LSTM) Cells

Specialized LSTM cells are incorporated to handle sequential data, preserving context over time and enhancing the model's ability to understand the temporal aspects of lip movements.

7. Temporal Feature Extraction:

The model extracts temporal features from the sequential data, contributing to a more comprehensive understanding of the temporal dynamics of lip motion.

8. Integration Layer:

An integration layer merges both spatial and temporal features, creating a unified representation that combines information from different aspects of the input data.

9. Fully Connected Layers:

These layers process the integrated features for classification, preparing the data for the final prediction stage.

10. Output (Text Prediction):

The final layer of the model predicts the corresponding text or phonemes based on the processed spatial and temporal features.

Implementation Plan

5.1 Schedule(Gantt Chart)

GANTT CHART

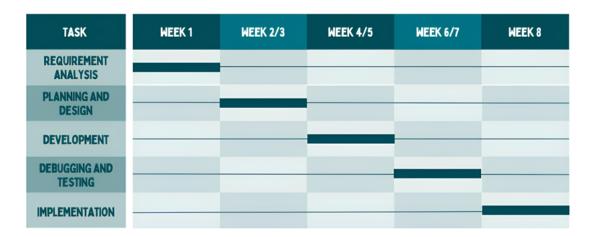


Figure 5.1: Gantt Chart

5.2 Hardware and Software Requirements

5.2.1 Software Requirements

5.2.1.1 Python and Deep Learning

We will be using Python as our programming language for this project. Python is a high-level general-purpose computer programming language often used to build websites and software, automate tasks, and conduct data analysis. It is simple, free, easy to use and highly compatible language consisting of a lot of libraries as well as built-in data structures. Having better library ecosystem, better visualization options, platform independence, and it's well known simplicity, consistency and flexibility, Python has proven itself to be one of the best picks for Artificial Intelligence and Machine Learning. Machine learning is a branch of Artificial Intelligence, where we start with an image and extract it's salient features. Then we create a model that describes or predicts the object on the basis of those features. On the other hand, for Deep Learning, we skip the manual step of extracting the features from the object and directly feed the images into a Deep Learning Algorithm, which then predicts the object. Deep Learning can be used to eliminate the limitations of Machine Learning since it makes it easier to handle complex problems as well as helps us predict through huge amount of data with ease too. Thus, Deep learning is a subset of machine learning which provides the ability to machine to perform human-like tasks without human involvement. It provides the ability to an AI agent to mimic the human brain. Deep learning can use both supervised and unsupervised learning to train an AI agent. Here we will try to utilize technique of Deep Learning and concepts of computerized neural networks using Python for the completion for this project. It serves as the primary programming language for lip reading project, providing a flexible and easy-to-read syntax.

5.2.1.2 OpenCV

OpenCV is an open-source computer vision library that provides a wide range of tools and functions for image and video processing. In our lip-reading project, we use OpenCV to capture and process video frames, apply image preprocessing techniques (such as resizing, filtering, and normalization), and extract relevant features from lip movements such as color or shape information. It also converts the preprocessed frames into a format suitable for input to a PyTorch model.

5.2.1.3 PyTorch

PyTorch is a deep learning framework that is widely used for building and training neural networks. In the context of lip reading, PyTorch can be employed to create and train deep learning models, such as convolutional neural networks (CNNs) or recurrent neural networks (RNNs). These models can learn to recognize patterns in lip movements and extract meaningful information for lip reading.

5.2.1.4 Matplotlib

Matplotlib is a plotting library for Python that allows to create a variety of static, animated, and interactive visualizations. In lip reading project, Matplotlib can be used for visualizing different aspects of data and results. For example, we use it to plot training/validation curves, display video frames with overlaid predictions, or create graphs to illustrate the performance of the lip reading model. The Matplotlib is used to visualize the training/validation loss curves during model training. By combining these tools, we can create a comprehensive lip reading system that leverage computer vision, deep learning and visualization to understand and interpret lip movements from video data.

5.3 Functional and Non-Functional Requirements

5.3.1 Functional Requirements

1. Pre-processing:

- Identify and track the face in the video sequence
- Extract the region of interest (ROI) containing the mouth.
- Normalize the ROI with respect to size and orientation.

2. Feature Extraction:

- Extract relevant features from the mouth region, such as lip shape, contour, and movement dynamics.
- Utilize deep learning models (e.g., Convolutional Neural Networks) to achieve accurate feature extraction.

3. Phoneme Recognition

- Based on the extracted features, classify the visual information into corresponding phonemes.
- Employ deep learning models trained on large lip-to-phoneme datasets.

4. Sentence Formation

- Combine the recognized phonemes into complete words and sentences using language models.
- Consider contextual information to resolve ambiguities and improve accuracy.

5.3.2 Non-Functional Requirements

1. Accuracy

- The system should achieve a high level of accuracy in translating lip movements to phonemes and subsequently to words and sentences.
- Specify a target accuracy percentage based on existing benchmarks or project goals.

2. Real-time performance

- The system should process and translate visual information with minimal latency, ideally in real-time.
- Define an acceptable delay threshold for lip-to-text conversion.

3. Robustness

- The system should perform well under varying conditions, including different lighting, facial expressions, and speakers.
- Specify the range of scenarios you want the system to handle efficiently.

4. User Interface

- The system should have a user-friendly interface for capturing video, displaying results, and interacting with the system.
- The system should have a user-friendly interface for capturing video, displaying results, and interacting with the system.

5. Resource Efficiency

- The system should be able to run efficiently on available hardware resources, without excessive memory or processing power requirements.
- Optimize the model and algorithms to minimize resource utilization without compromising accuracy.

Expected Outcomes

The Lipreading algorithm is expected to speech recognition better by usage of especially in noisy places. Anticipated outcomes involve achieving enhanced accuracy through the integration of cutting-edge deep learning architectures and ensemble learning techniques. Furthermore, the objective is to help people with hearing impairments and improve the accuracy in audiovisual applications.

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