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# LIPS: Let's think of a better title...

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## Abstract

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## 1 Introduction

Lip reading is the process of identifying spoken words purely from lip movement without any audio data. It is notoriously difficult, even for humans, but it plays a crucial role in human communication. This is highlighted by the McGurk Effect, where one phoneme's audio is transplanted on top of another phoneme's video, causing a third phoneme to be perceived [1]. Hearing-impaired people especially rely on this method to understand speech without audio. Even the most skilled lip readers are able to achieve an accuracy of around 30% when identifying isolated words. They rely on context, body cues, and hand gestures to fill in the gaps [2]. Commonly, humans also use lip reading in cases where discreet communication is necessary or in noisy environments where audio signal is not available.

With recent machine learning advances, attention has been drawn to this problem due to its many practical applications. Lip reading has promising applications in assisting hearing-impaired people. With automated lip reading, video annotation and translation become possible. Lip reading also has applications in surveillance and analysis of forensic video.

In this paper, we propose LIPS, a method for lip reading via deep learning. Our method treats lip reading as a word classification problem. After extensive video processing, we use convolutional neural networks to exploit spatio-temporal features from input video data.

## 2 Related work

In recent years, there have been many advances in lip reading. Preceding work can be organized into two groups: classification systems and sequence prediction systems.

### 2.1 Classification Systems

The majority of work done on this problem has been in the development of classification systems. The classification system approach performs classification over words or phonemes rather than full sentence sequence prediction. Ninomiya et.al. used Hidden Markov Models for classification of words or phonemes [3]. Sui et.al. approach the lip reading problem with Boltzmann Machines [4] and Chung et.al. propose a deep neural network framework for word classification. Their method can effectively learn and recognize hundreds of words [5]. Thabet et. al. propose novel video processing methods and show experimental results over multiple classifiers including Multi-Layer Perceptron,

Naive Bayes, Support Vector Machine and Logistic Regression. Their results were very impressive with accuracy of 65% over 5 distinct words [6].

## 2.2 Sequence Prediction Systems

Comparatively, much less work has been done on sequence prediction systems for lip reading. These systems aim to translate end-to-end sentences from visual data rather than single word classification. Notable work in this category was done by Graves et.al who use deep learning with recurrent networks for end-to-end speech recognition [7]. Their work, while not strictly lip reading, influenced following sequence prediction for lip reading such as the work of Assael et.al. with their system LipNet. LipNet uses convolutional neural networks to provide a distribution over sequences of phonemes and demonstrates high accuracy, especially with respect to distinguishing phonemes [8].

## 3 Method

Transforming video footage into a lip-reading prediction requires a significant processing pipeline. In this section, we discuss the training dataset, how it was prepared for modeling, and the overall architecture of our model.

### 3.1 Data

The Oxford-BBC Lip Reading Sentences 2 (LRS2) Dataset was selected for training. Details of its creation can be found in [9]. A brief summary of the data will follow.

The atomic units of raw data are video files, each around ten seconds long. The video clips come from BBC news and talk shows, but each frame's viewing window is shrunk to track the face of an individual speaker. A single speaker may span multiple clips, but LRS2 was designed specifically to include as many different faces as possible. Accompanying each video is a like-named metadata text file; this file contains an ordered list of words spoken in the clip along with timestamps indicating when the speaker starts and stops saying each word. These times are precise to the hundredth of a second.

We did not use the entirety of LRS2; we limited our study to the "pre-train" segment because the "main" segment's text files do not include word timestamps. Still, the model's potential training examples encompassed over 2 million utterances of 41,427 unique words; 47GB of data in all.

### 3.2 Feature Processing

Consider an example pipeline whose output is a model that, given a video clip, predicts whether the speaker in that clip is saying the word "yes" or the word "no." So as to act within the realm of computational feasibility, we assign a parameter  $N_{files}$  as an approximator for the size of this model's training set.

First, we prepare a dictionary mapping strings to unique integers. We formally define this as a set  $D = \{\text{'YES'} : 0, \text{'NO'} : 1\}$ . Next,  $N_{files}$  text files are parsed word by word. If a word in the file is 'YES' or 'NO,' it is recorded, along with its beginning and ending timestamps.

Next, the videos corresponding to those  $N_{files}$  text files are loaded and converted to greyscale. If a clip contains a word belonging to  $D$ , every frame between that word's two timestamps is recorded into its own 2D array of integers between 0 and 255. Thankfully, the videos in LRS2 are already of a uniform height and width, so no resizing is necessary. Together, this list of 2D arrays composes a 3D object we call a **word clip**. Note that at this stage, audio information is no longer present in the data representing the video.

Of course, each word instance may be composed of a different number of frames, depending on how long the speaker took to say the word. The model, however, will expect every input to be of the same dimension. Thus, we find the median length in frames of every word clip already created. If a word clip's frame count is higher than the median, it is evenly downsampled. Word clips that are too short have some of their frames repeated until they reach the median frame count. At the end of this preliminary process, then, we have a list of data objects pairing a word with a 3D word clip object of uniform size.

Until now, we have dealt with word clips of the speaker’s entire face. However, the model’s task is far simpler if it only sees the speaker’s mouth. . . .

Thus, at the end of the pipeline, the model is prepared to be trained. The input vector is a flattened representation of a speaking mouth over some uniform number of frames; The target is a number that maps to some word in  $D$ .

To generalize this pipeline, we simply redefine  $D$  to include whatever words the model needs to distinguish between. To train models quickly, we can reduce  $N_{files}$  and temporarily store processed video data; to validate them, we can increase  $N_{files}$  to include the entire training set.

### 3.3 Model

## 4 Experimental Results

## 5 Conclusion

### Acknowledgments

Use unnumbered third level headings for the acknowledgments. All acknowledgments go at the end of the paper. Do not include acknowledgments in the anonymized submission, only in the final paper.

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