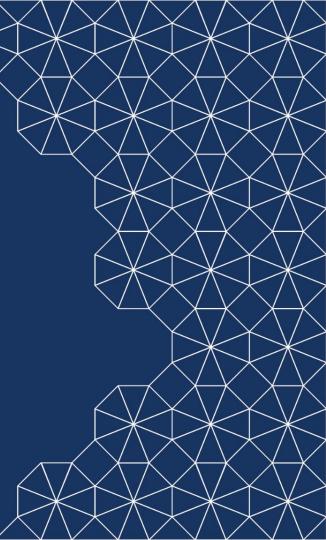
# Multi-modal Emotion Detection and Classification using Audio-Visual Data

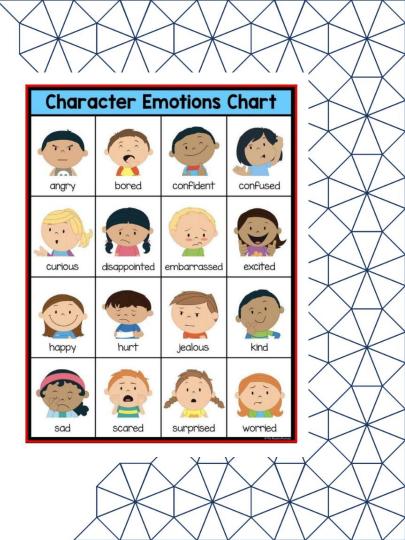
Violet Li, Roxy Rong, Weijie Yang





## **Problem statement**

This project aims to solve the **emotion classification problem** from both **visual** and **audio** data, without having the knowledge of the speaker information.



## **Dataset: SAVEE Database**

- The SAVEE database was recorded from 4 native English male speakers.
- There are **7 categories of emotions**: anger, disgust, fear, happiness, sadness, surprise and neutral.
- It contains videos of their facial expressions and audios of them reading emotion-specific sentences.











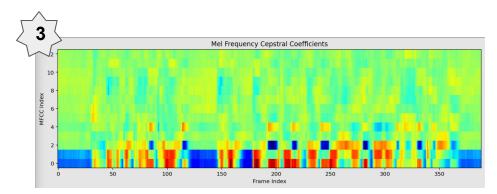


# Audio Experiments

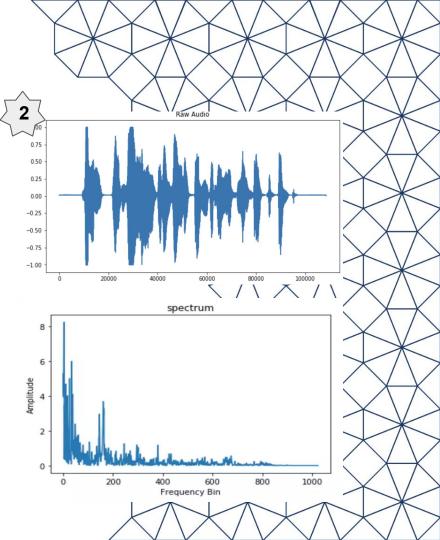


## **Audio Features Extraction**





\* Time period, frequency, amplitude



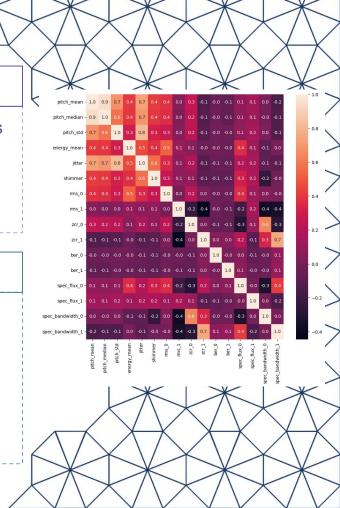
## **ML Models - Audio**

#### **Audio Feature Selection**

- Basic features + PCA on 1 dimensional array features -> 40+ features
- Correlation matrix to reduce it to 16 features
- All inputs are normalized

#### **Model Architecture**

- Traditional ML model: LogReg, SVM, RF, KNN
- CNN Model (MFCC -> 2 layers Conv2D, Kernel = (5, 15), (3, 9))
- Dual CNN Model & CNN + other features Model
- Use Early Stopping, Regularization, Kernel Constraints to prevent overfitting with more complicated models



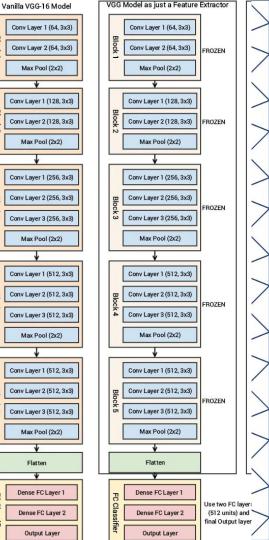
## **Transfer Learning**

#### **Why Transfer Learning**

- As we have a **small dataset**, transfer learning allow use start from a well-trained model.
- Reduce computational time & dataset needed dramatically
- Only train the new added dense layers

#### **Steps**

- Determine which base model to use.
- Preprocess the dataset so that it is a valid input.
- Train the model



Max Pool (2x2)

Flatten

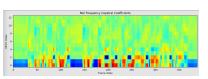
Dense FC Layer 1

Output Laver

Audio: Model based on VGG19 / YAMNet

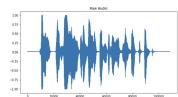
#### **MFCC Diagram with VGG19**

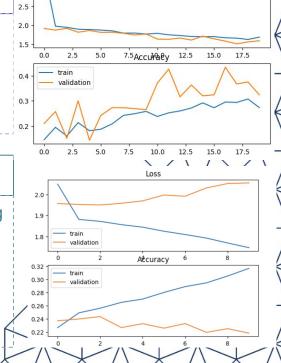
- convolutional neural network architecture consisting of 19 layers
- Take images as input.
- Test Accuracy: 0.39



#### **Waveform with YAMNet**

- A pre-trained neural network that employs the MobileNetV1, classifying audio types.
- Use audio waveform as input.
- Over-fitting as we increase epochs.
- Test Accuracy: 0.21





3.0

validation

# Why Transfer Learning is not working so well

- Though transfer learning needs fewer samples, 480 original samples are still too few to train a good model.
- Not ideal base model selection.
  - VGG19 model is more for classifying images
  - YAMNet for classifying types of phonating objects (cars, animals, etc.).
  - Our samples will likely all get 'speech' embeddings without obvious differentiation.

To improve the results

- Extend the dataset.
- Not only use models as embeddings extractor, but fine-tune some existing layers

print(f'The main sound is: {inferred\_class}')
print(f'The embeddings shape: {embeddings.shape}')

The main sound is: Speech
The embeddings shape: (10, 1024)

## Model Performance Evaluation Overview (audio only)

Model	Accuracy	Precision	Recall	f1_score
Log Reg	0.385	0.37	0.38	0.37
SVM	0.447	0.43	0.45	0.44
RF	0.464	0.43	0.46	0.42
KNN (MFCC)	0.397	0.37	0.40	0.38
CNN	0.505	0.49	0.50	0.49
Dual CNN	0.422	0.48	0.42	0.44
CNN + others	0.552	0.53	0.55	0.53
Transfer Learning (VGG19)	0.389	0.35	0.39	0.39
Transfer Learning (YAMNet)	0.210	0.11	0.21	0.10

## **Limitations on Audio**

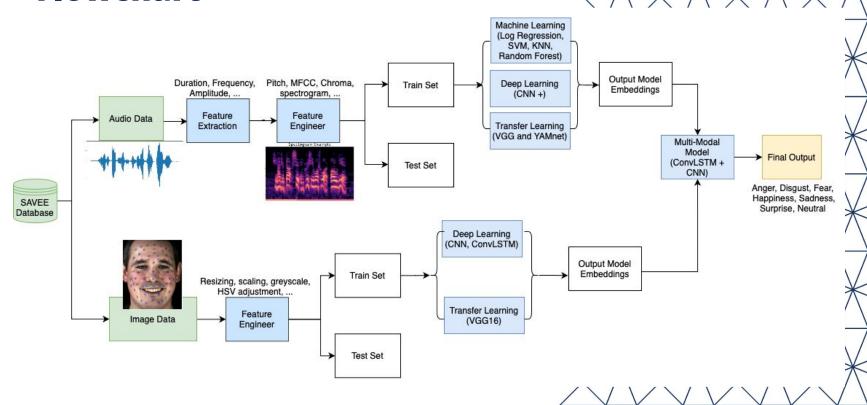
#### Weakness:

- Audio Model can't separate disgust or sad from neutral well, also can't distinguish between fear and surprise well
- Overall performance is still too low, not achieving 80% accuracy goal

#### **Next Steps:**

- Add visual data
- Build multimodal fusion

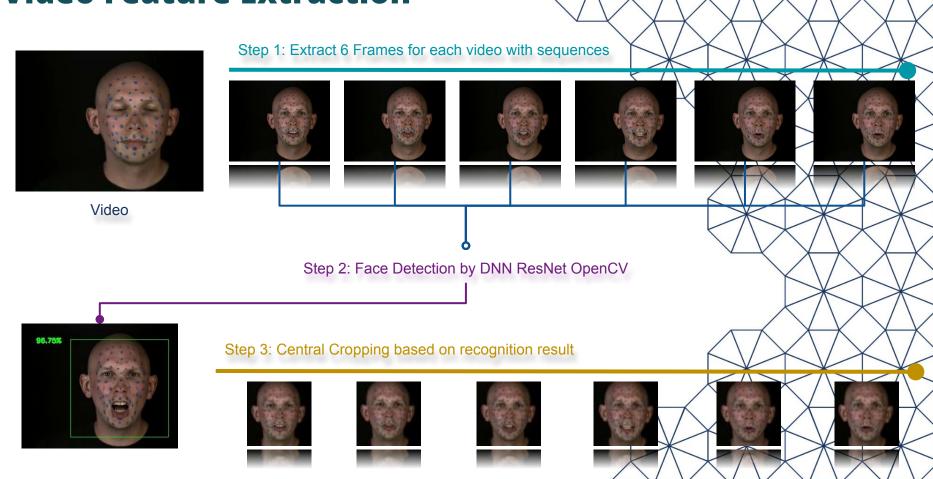
## **Flowchart**



# Image Experiments



## **Video Feature Extraction**



## **Image: Model based on VGG16**

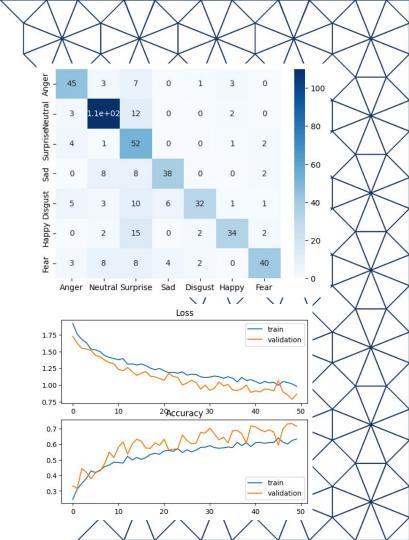
#### VGG16:

- convolutional neural network architecture consisting of 16 layers
- Trained on ImageNet, which contains more than 14 million training images across 1000 object classes.

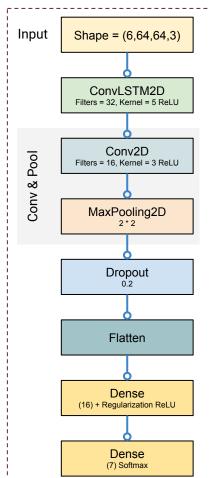
Test Accuracy achieved after 50 epochs: 0.73

#### Next Steps:

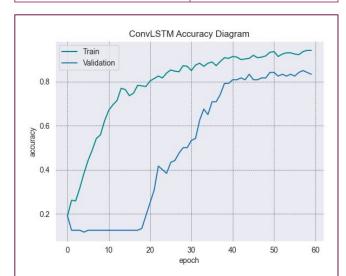
 Build a model that can make use of the temporal information

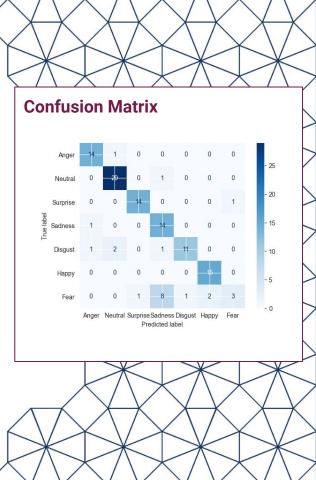


## **ConvLSTM on Visual Inputs**



Training Result					
Training Accuracy	0.9417				
Test Accuracy	0.8333				
precision	0.84				
recall	0.81				
f1-score	0.79				





## Multimodal Fusion

Concatenate both video and audio extractors into Neural Networks.

## **Model Input Processing**

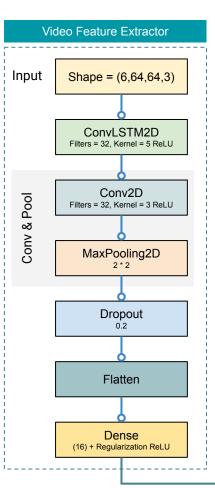
#### **Video & Audio Feature Processing**

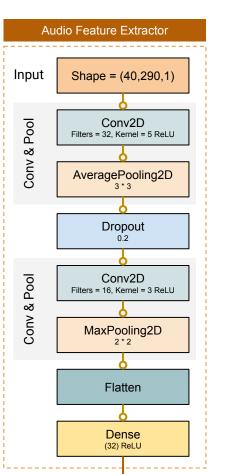
- Video Input Size: (6, 64, 64, 3) <6 frames with 64\*64 pixels RGB images>
- Audio Input size: (40, 290, 1) <MFCC with 40\*290 pixels non-RGB images>
- All inputs are normalized

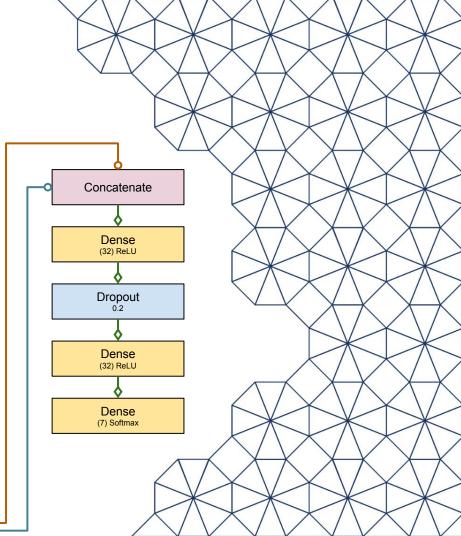
#### Train/Test Split Technique

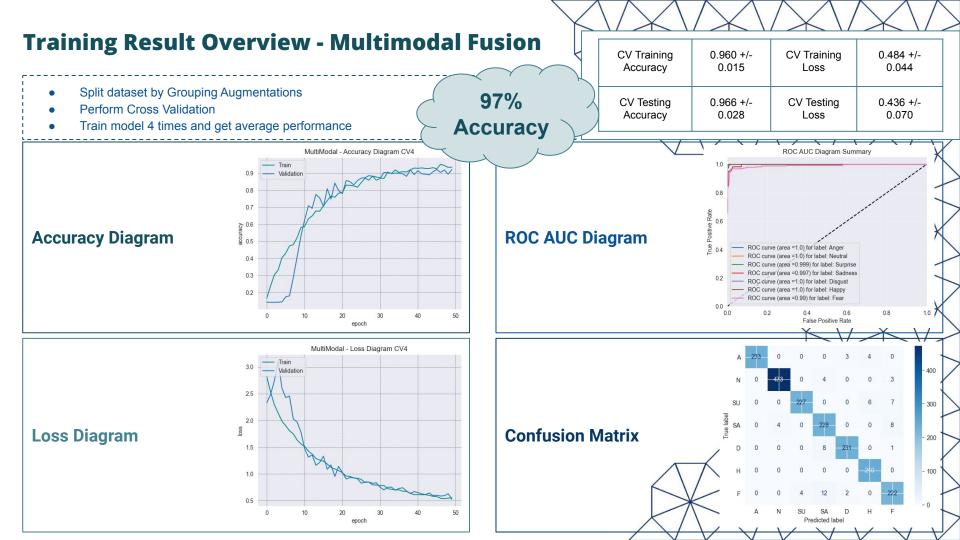
- Audio Dataset is augmented → Split the dataset by group
- Match video input with audio input
- Balance the portion of different emotion inputs → Weights Setting
- Cross Validation → Stratified Group K Fold (K = 4)

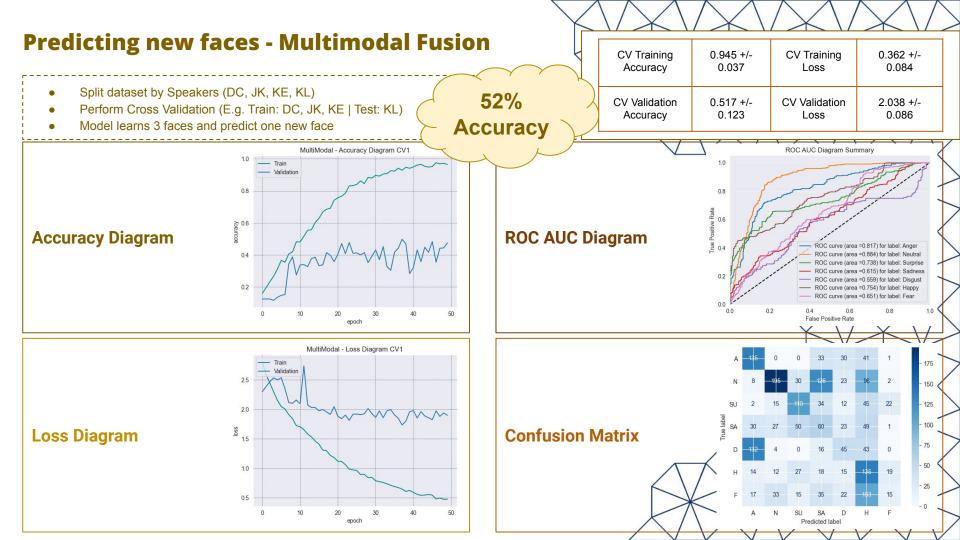
## **MultiModal Architecture**













## **Limitations (1-2mins)**

- Dataset Size: Not enough video data (only 4 faces, and all white male) make it harder to generalize
- Tuning: More complicated model structure makes hyper-param tuning much harder
- Al Fairness: All speaker are white males. This means that our sample contains bias on age, race, gender, and education level.

## **Discussion & Conclusion**

#### Result

- ConvLSTM Multi-modal fusion has the best performance.
- Potential reason: Emotion expression is a temporal behavior with visual and audio outputs. Multi-Modal can better capture temporal info and both features.

#### Reflections

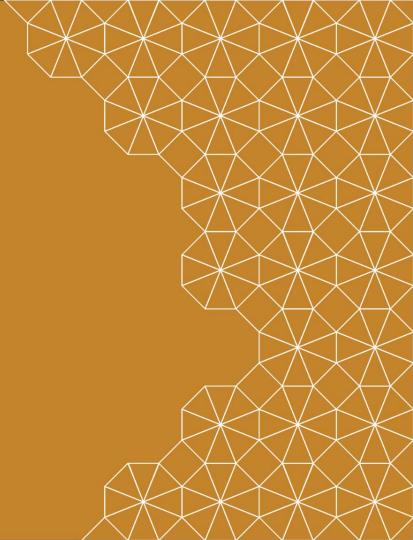
Sample selections

#### **Future Work**

- Add more samples and be more inclusive

# **Appendix**





#### A list of features:

teo

f0 pitch\_mean, pitch\_median, pitch\_std, pitch\_range, pitch\_max energy\_mean, energy\_median, energy\_std, energy\_range, energy\_max jitter, shimmer amplitude\_envelope rms zcr ber amplitude\_spectrogram dB\_spectrogram spec\_contrast spec\_flux spec\_centroid spec\_bandwidth spec\_flatness power\_spectrogram mel-spectrogram mfcc mfcc\_delta

## **Data Augmentation**

• Only 480 samples, so we need to augment the dataset.

Original file

- Methods used:
  - a. Add Gaussian noise 🕩
    - b. Pitch Scaling
    - c. Time Stretching
    - d. Random Gain
    - e. Invert Polarity
- Randomly combine these methods to obtain augmented dataset.

### **Performance Overview - Multimodal Fusion**

