



國立臺灣大學

National Taiwan University

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# Mockingjay: Unsupervised Speech Representation Learning with Deep Bidirectional Transformer Encoders

Andy T. Liu, Shu-wen Yang, Po-Han Chi, Po-chun Hsu, Hung-yi Lee

Speech Lab, National Taiwan University (NTU)

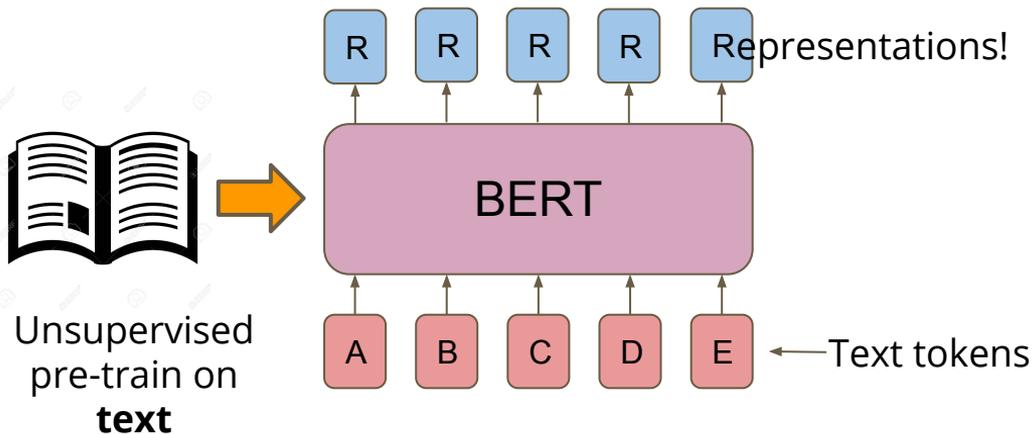
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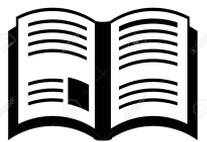
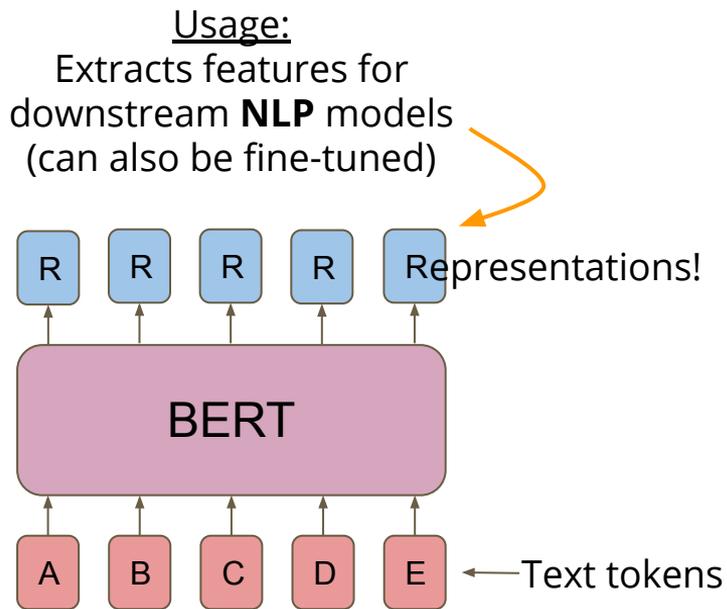
# Introduction

## NLP BERT: Language Representation Learning



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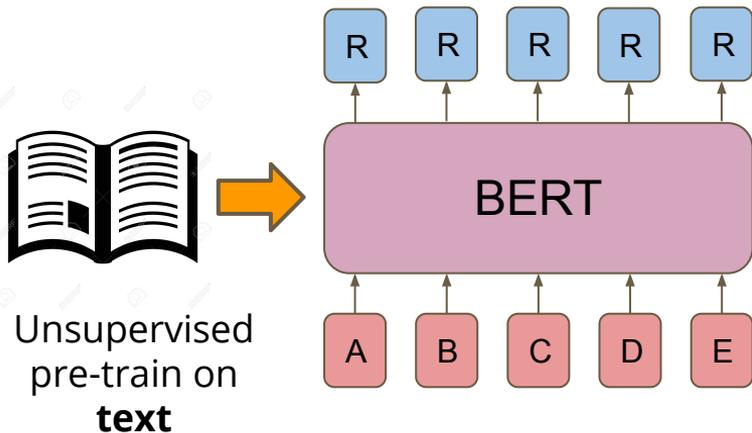


Unsupervised  
pre-train on  
**text**

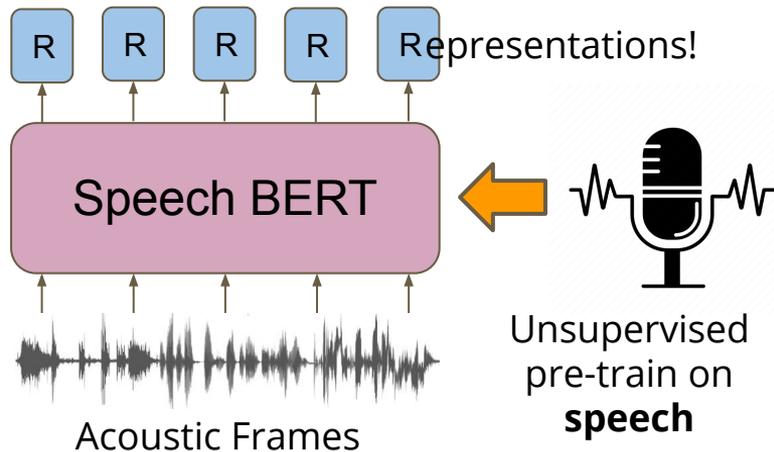
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## NLP BERT: Language Representation Learning

Usage:  
Extracts features for  
downstream **NLP** models  
(can also be fine-tuned)



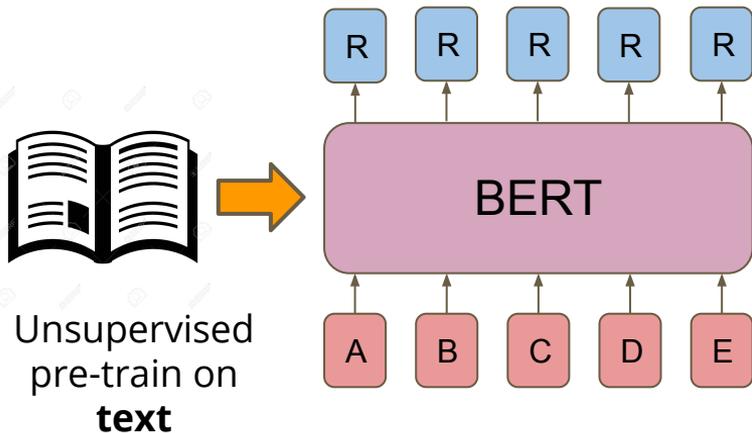
## Speech BERT: Speech Representation Learning



# Introduction

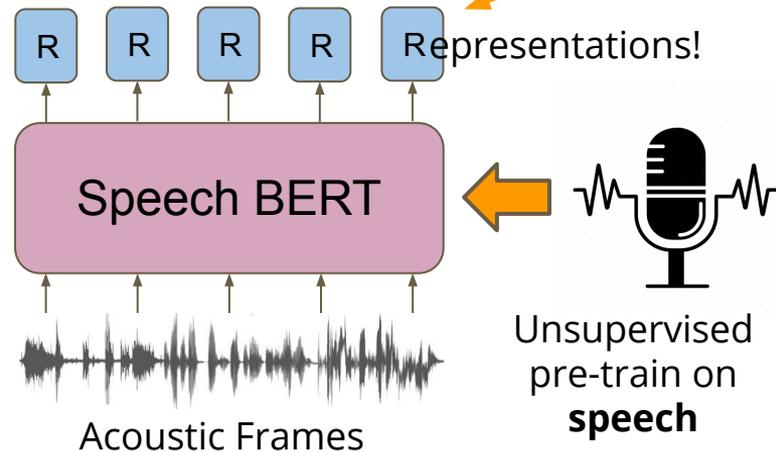
## NLP BERT: Language Representation Learning

Usage:  
Extracts features for  
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(can also be fine-tuned)



## Speech BERT: Speech Representation Learning

Usage:  
Extracts features for  
downstream **SLP** models  
(can also be fine-tuned)



# A View of Recent Unsupervised Speech Representation Learning Approaches

July, 2018  
DeepMind



**CPC**

Phone / Speaker

[1] Representation Learning with Contrastive Predictive Coding

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[1] Representation Learning with Contrastive Predictive Coding

↑  
Compares with

April, 2019  
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**APC**

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[2] An Unsupervised Autoregressive Model for Speech Representation Learning



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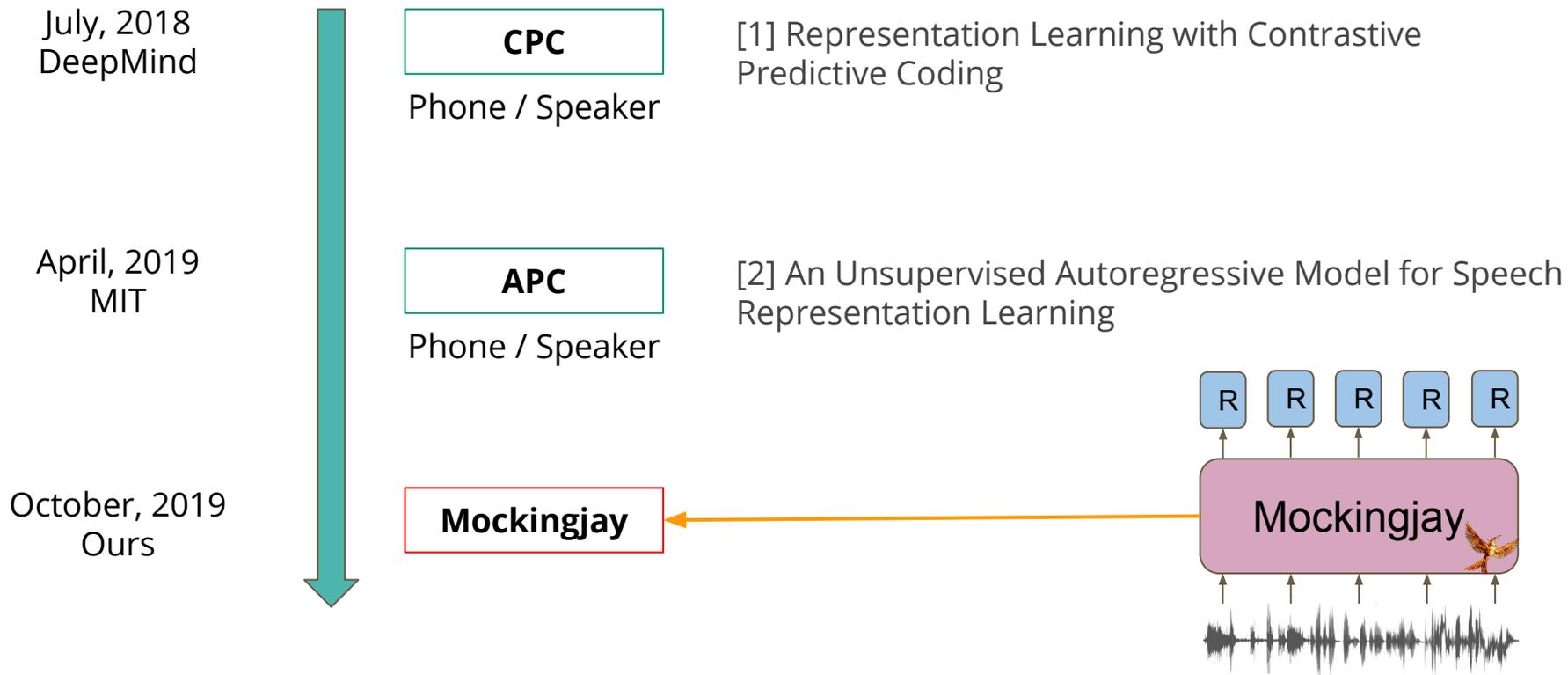
[2] An Unsupervised Autoregressive Model for Speech Representation Learning

## Common heuristic:

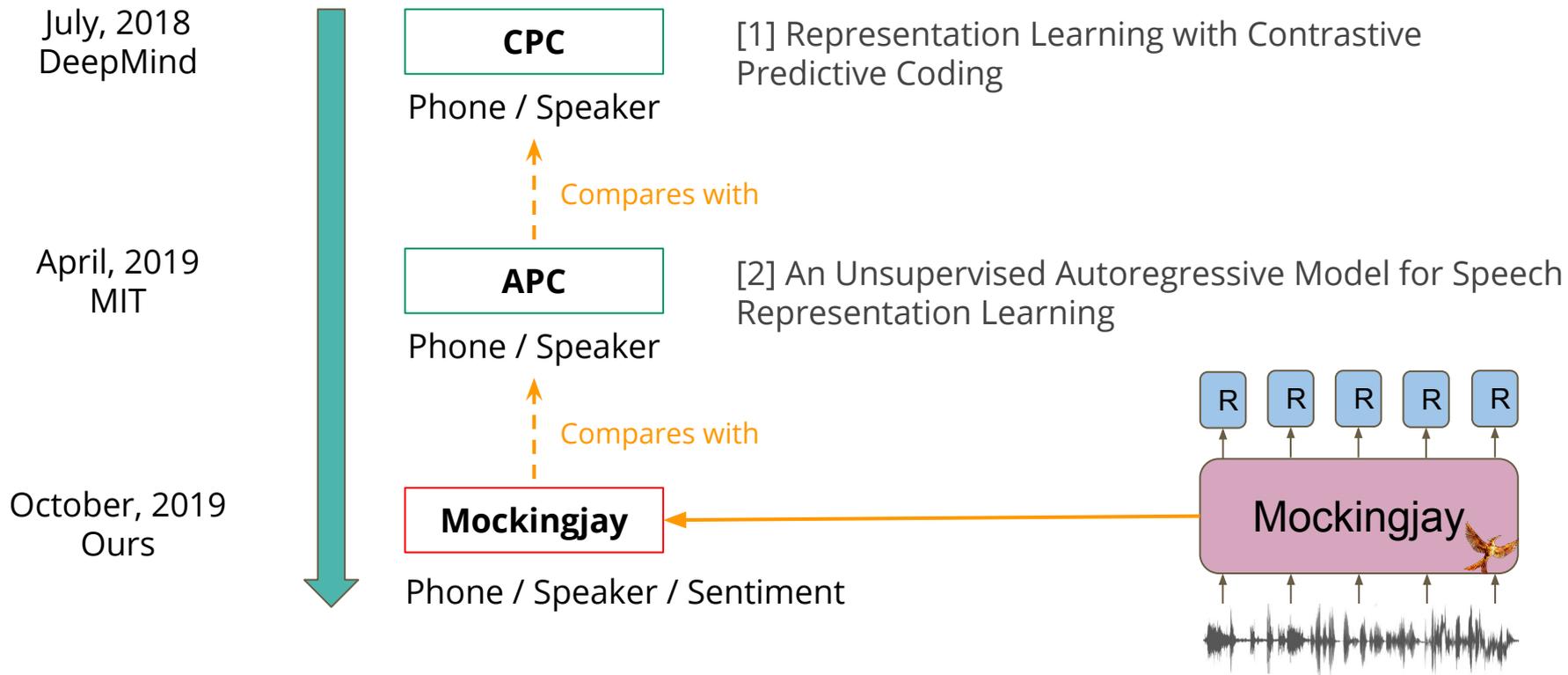
They both encode past information and predict information about future frames.



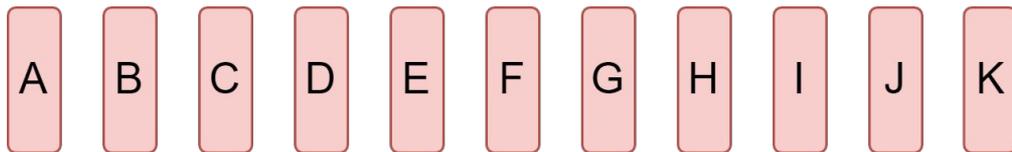
# A View of Recent Unsupervised Speech Representation Learning Approaches



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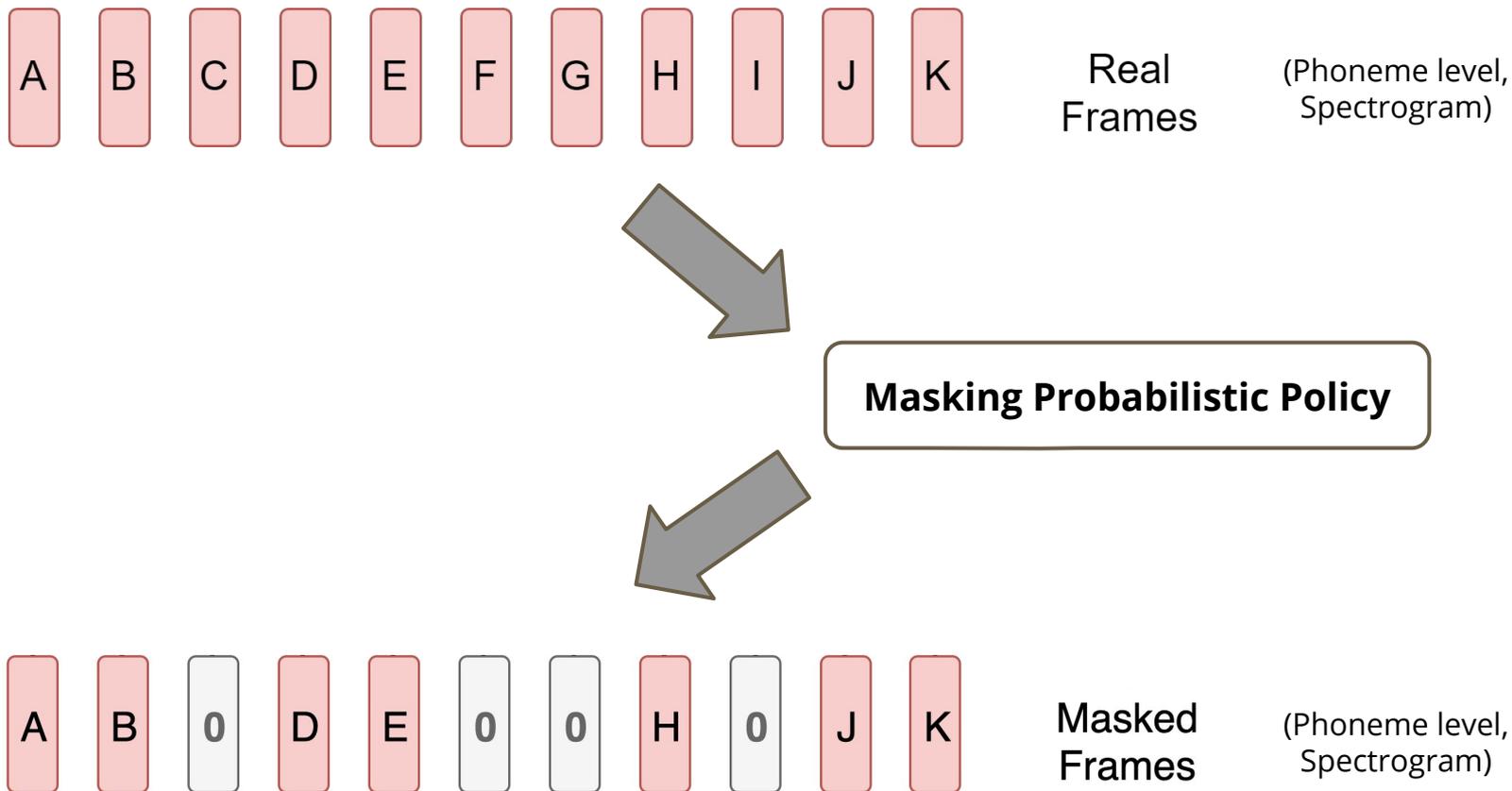
# Pre-Training Task: Masked Acoustic Model



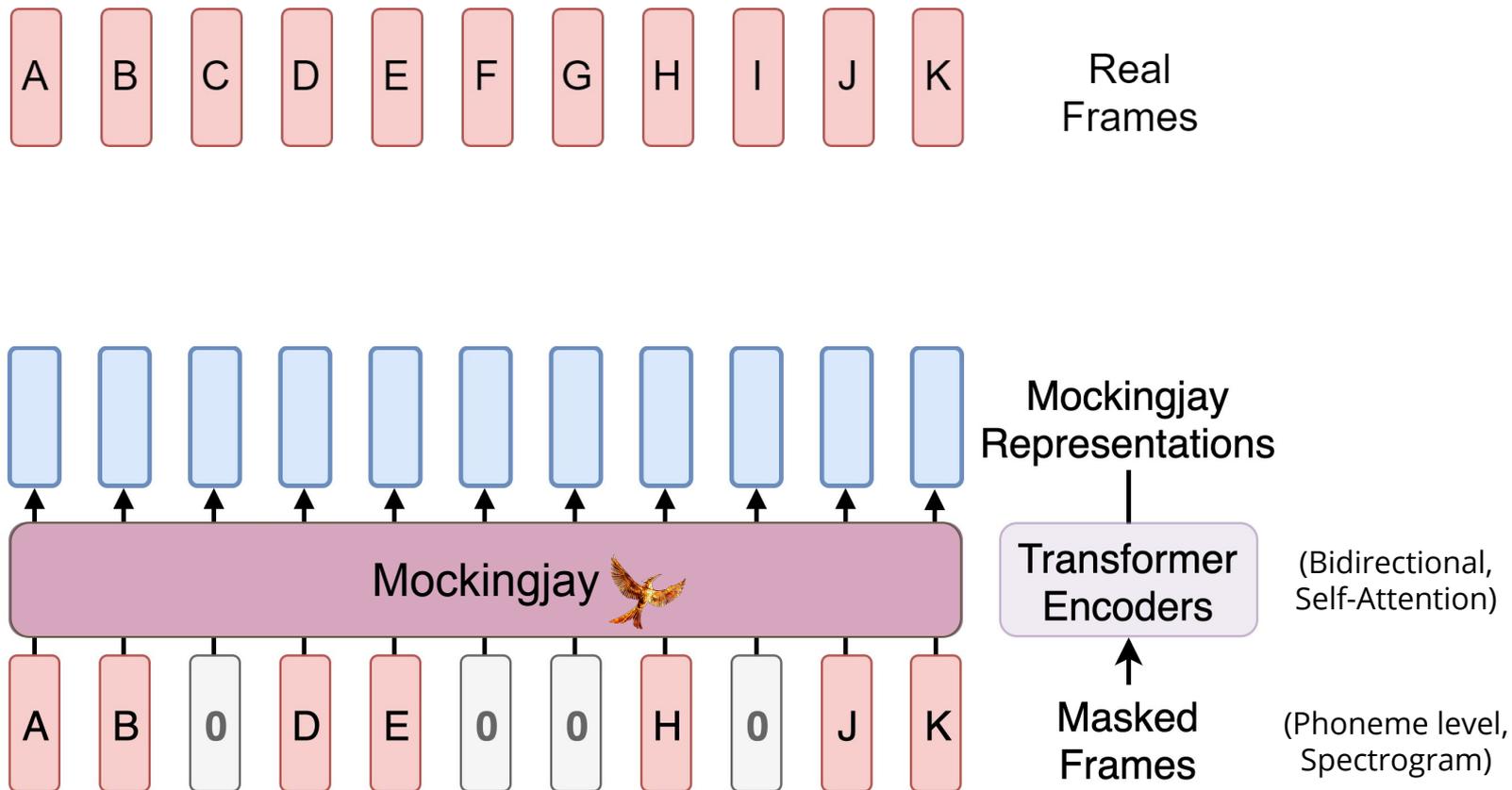
Real  
Frames

(Phoneme level,  
Spectrogram)

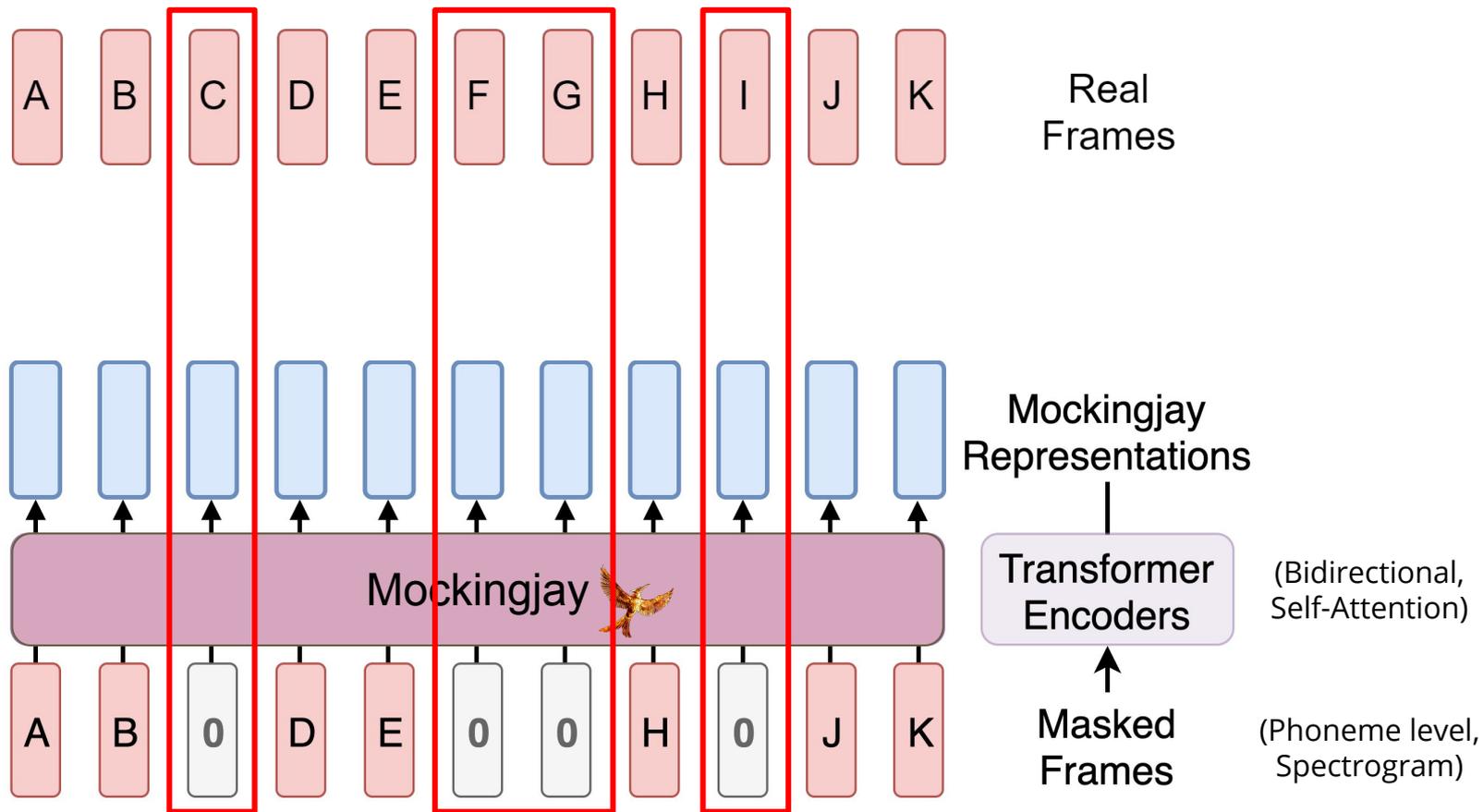
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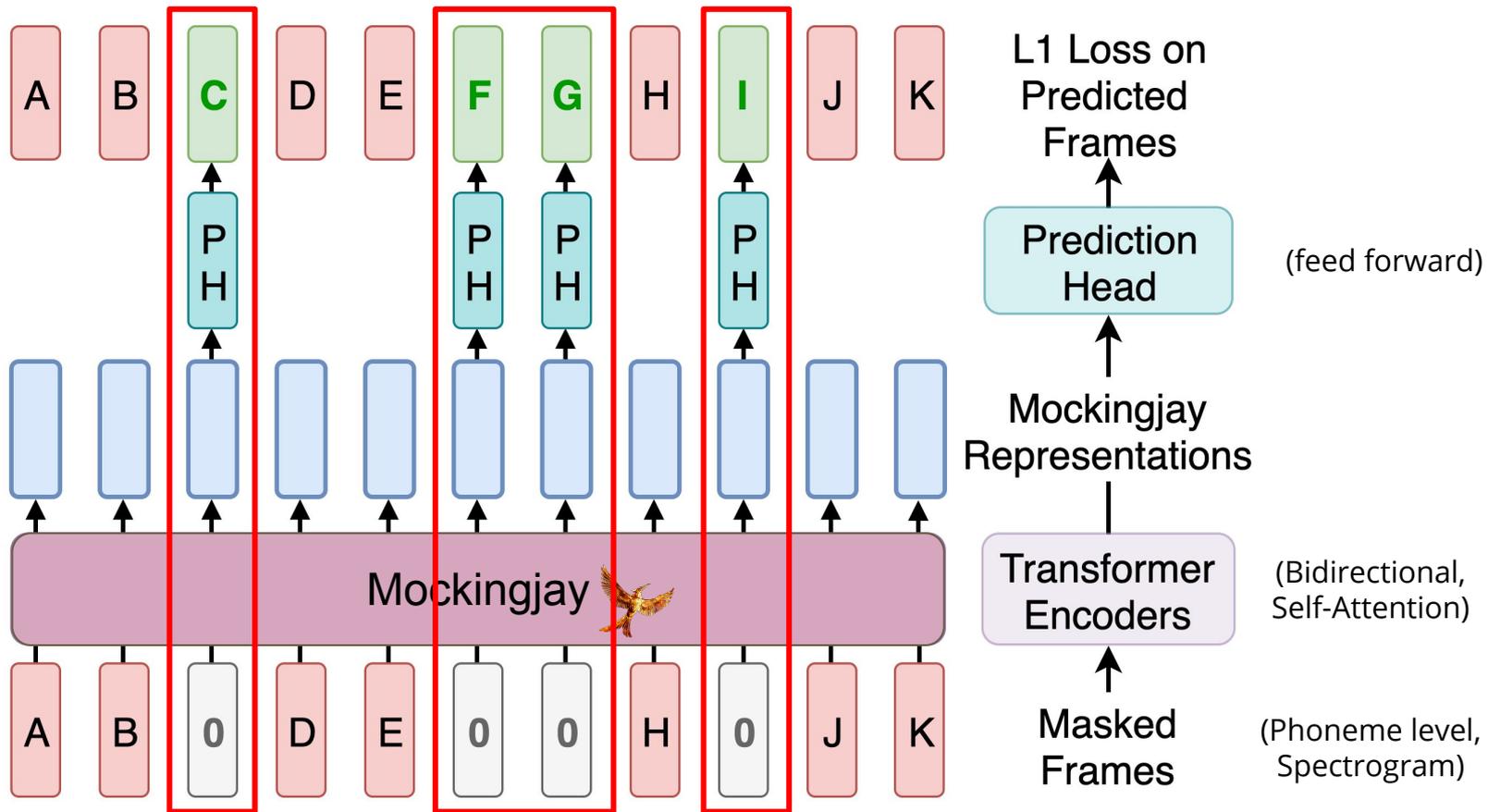
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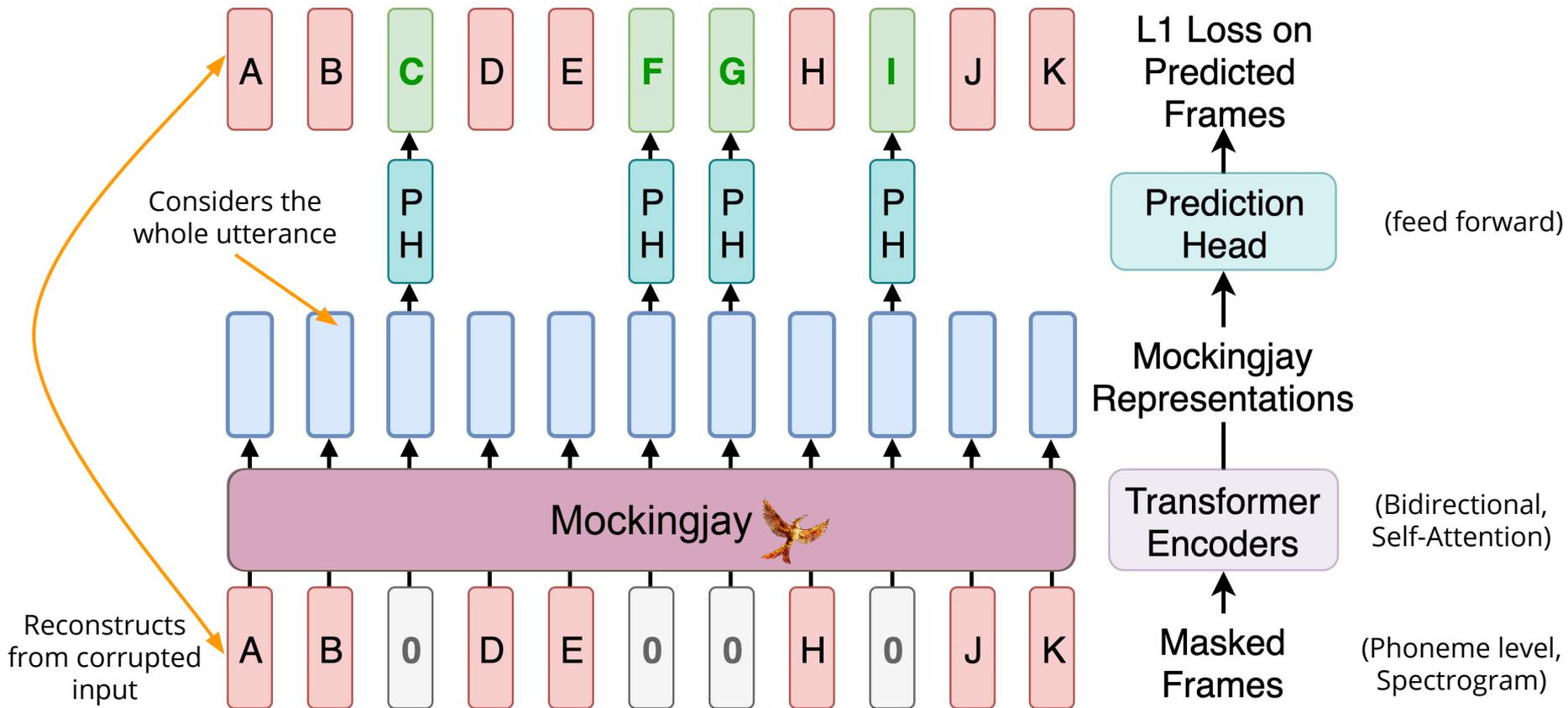
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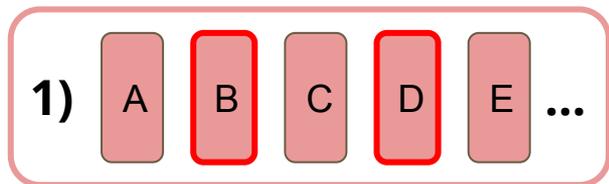
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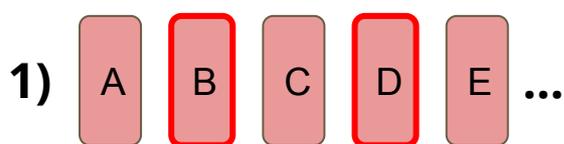


# Probabilistic Policy for Masking Frames

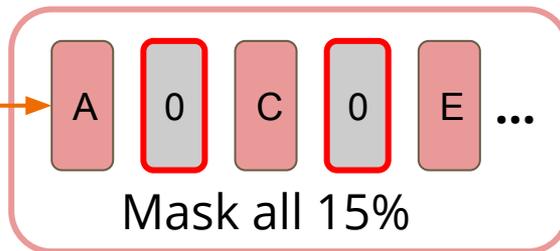


1) Select **15%** of the frames for prediction (highlighted in green).

# Probabilistic Policy for Masking Frames



2) 80%

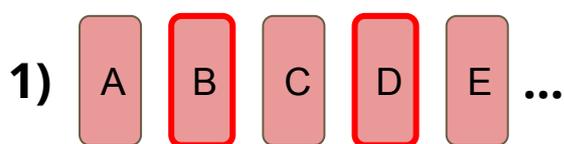


1) Select **15%** of the frames for prediction (highlighted in green).

2) For all selected frames:

- mask to zero **80%** of the time
- replace randomly **10%** of the time
- leave untouched **10%** of the time

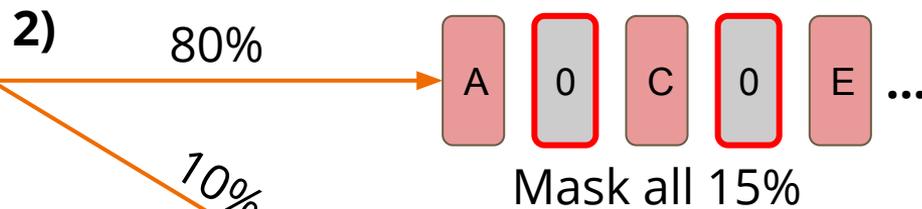
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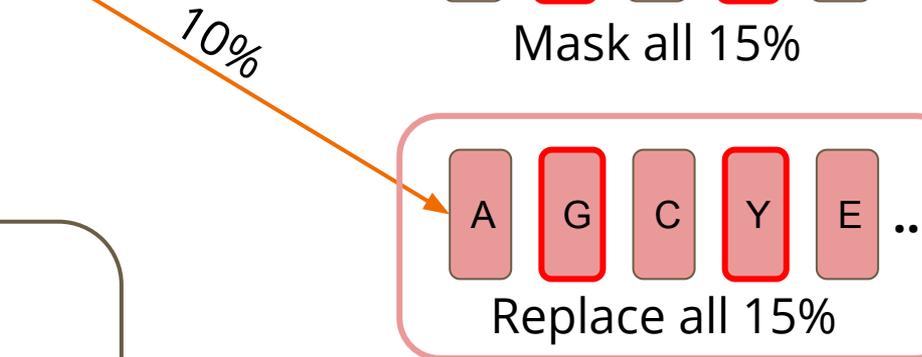
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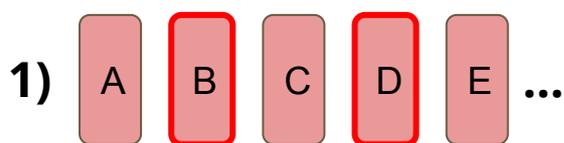


Mask all 15%



Replace all 15%

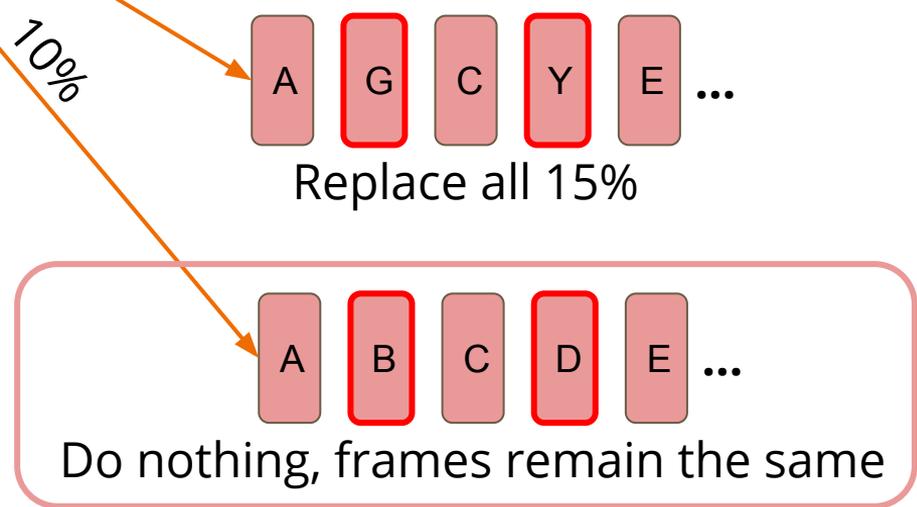
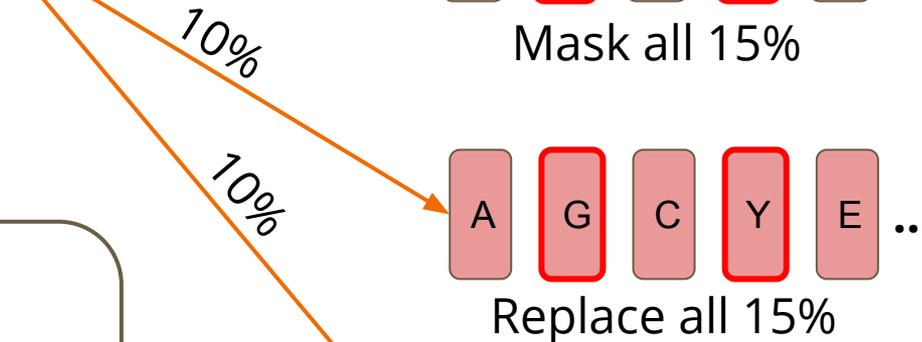
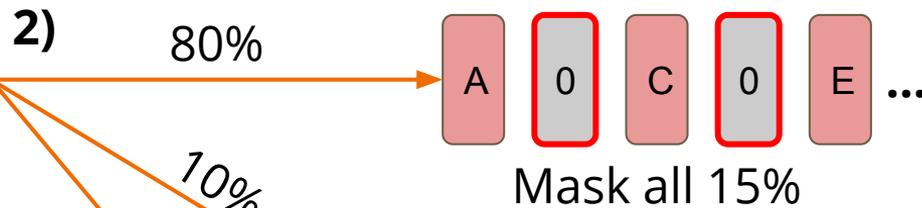
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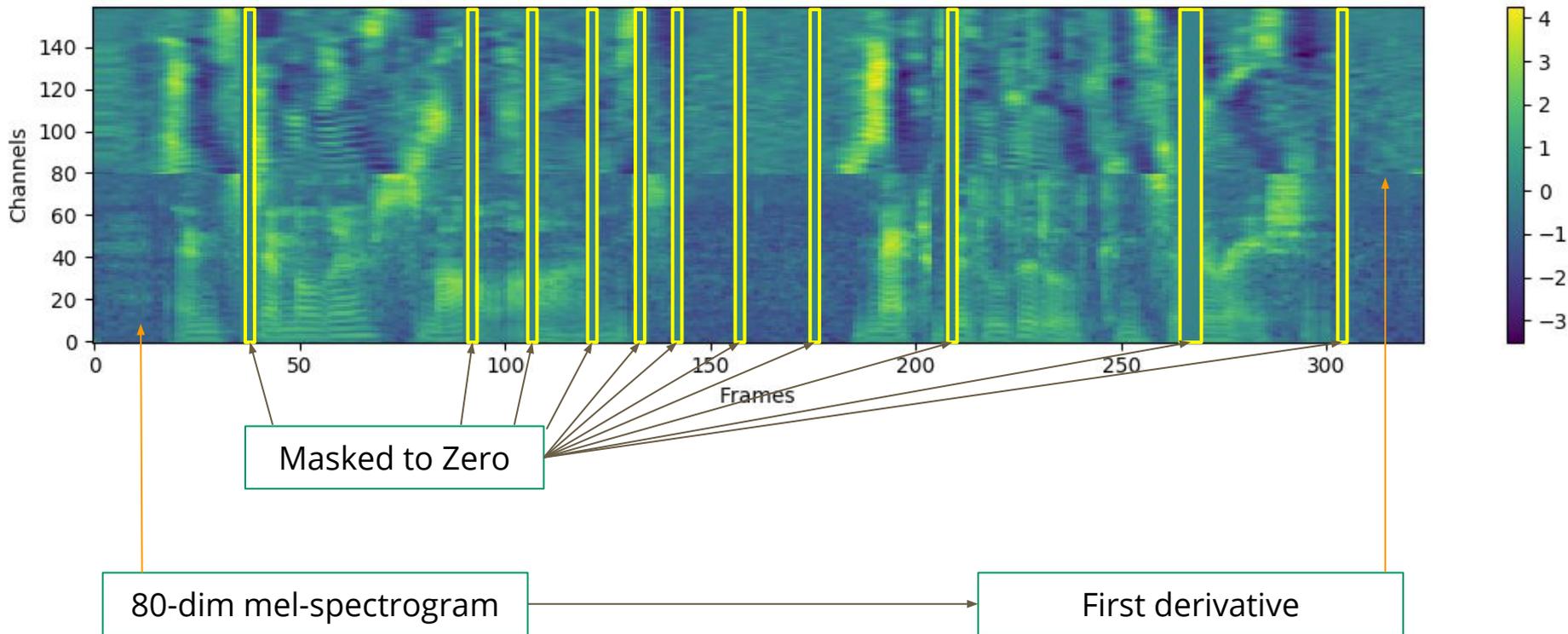
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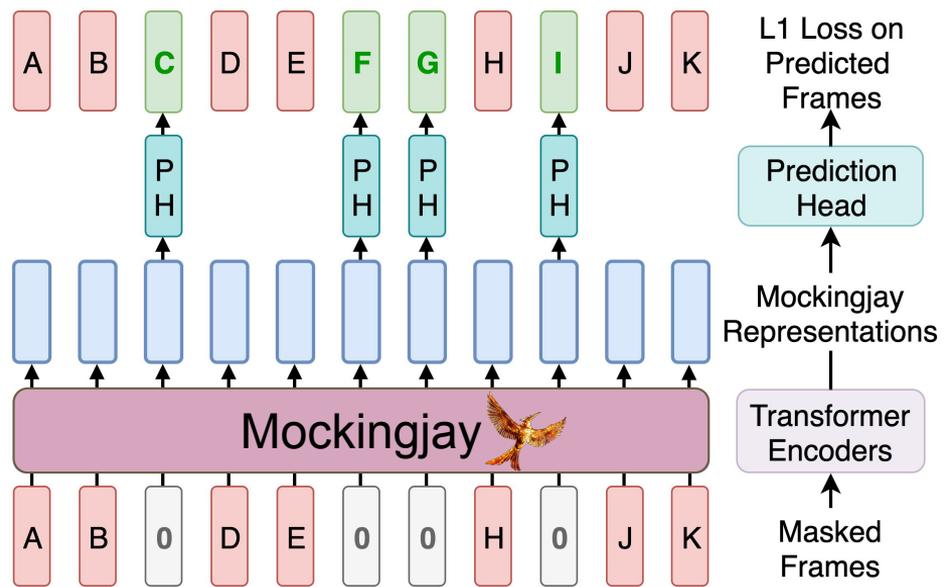
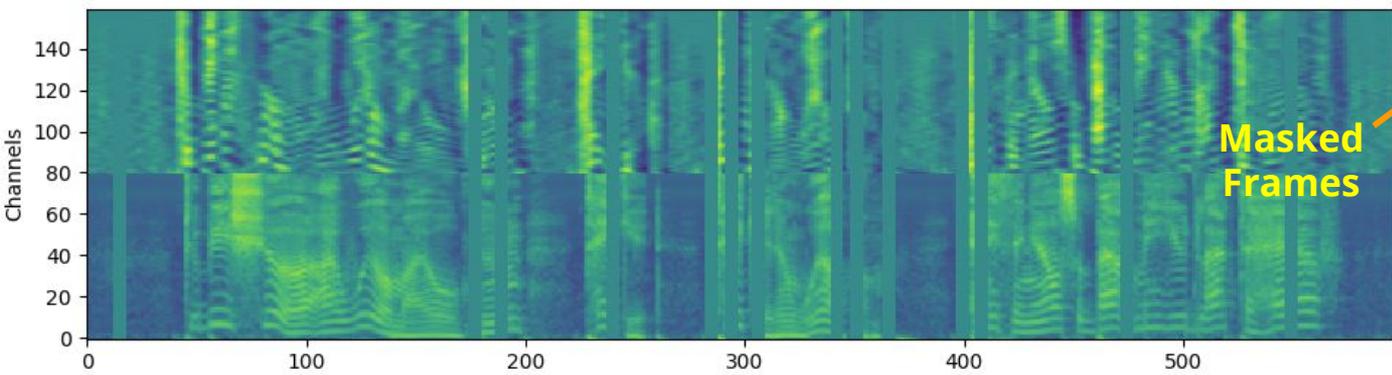
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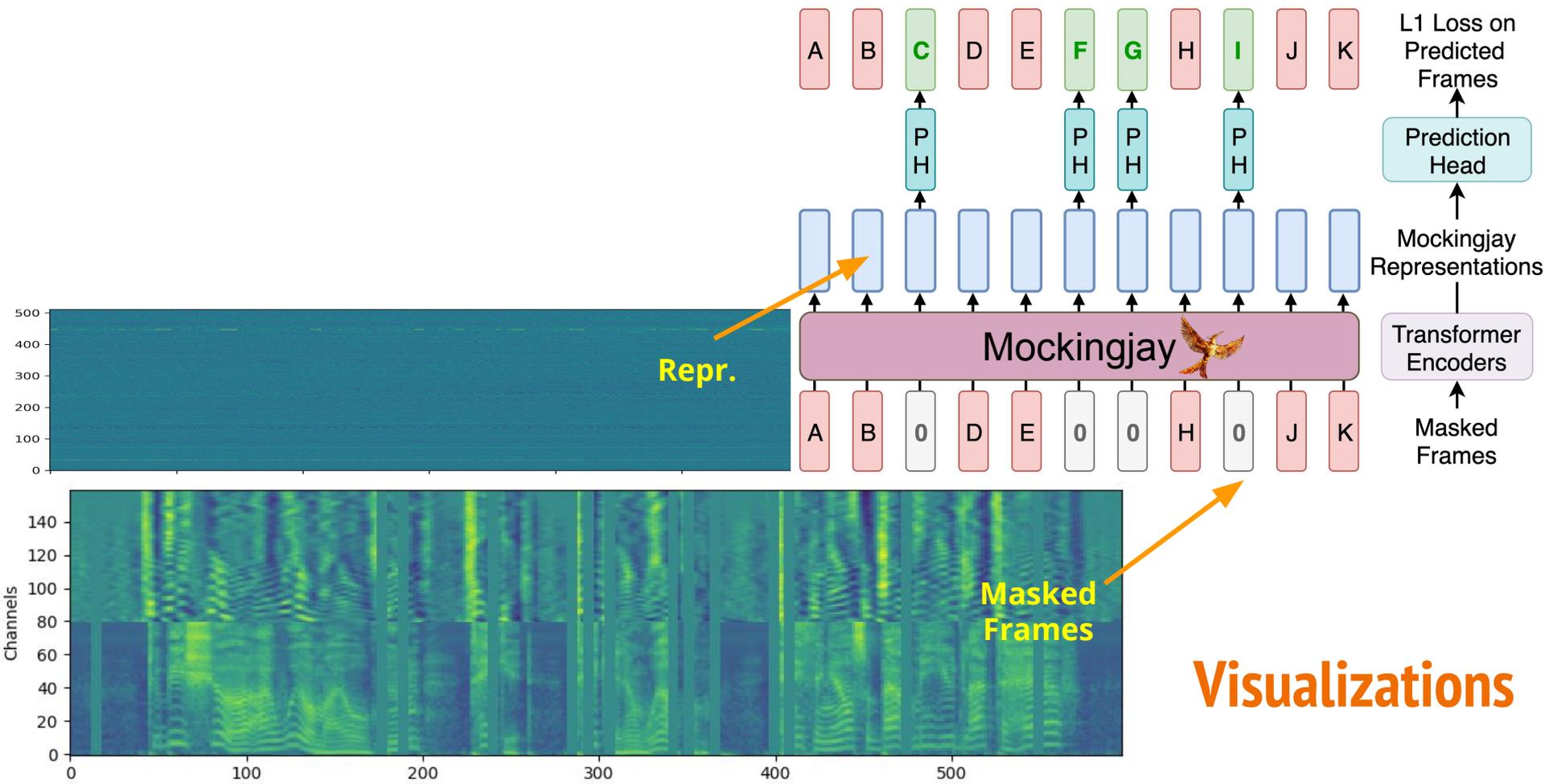


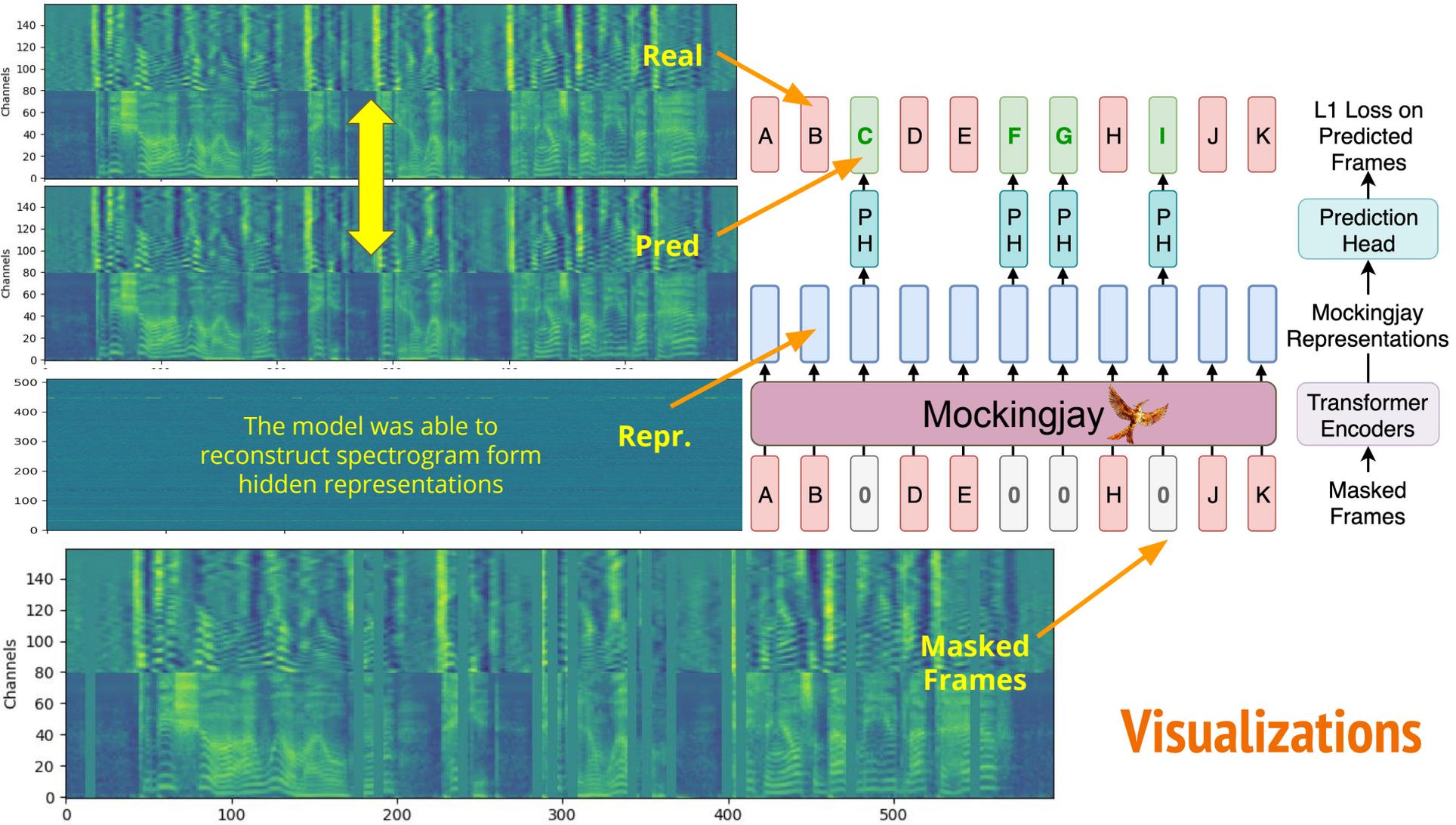
# Input Feature: Masked Spectrogram





**Visualizations**





# Migrating from text to speech

**Acoustic Features:** long and locally smooth in nature,

need to 1) shorten the sequence and 2) mask over a longer span



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Address the long and smooth problem with:

*Downsampling*, and *consecutive masking*

# Migrating from text to speech

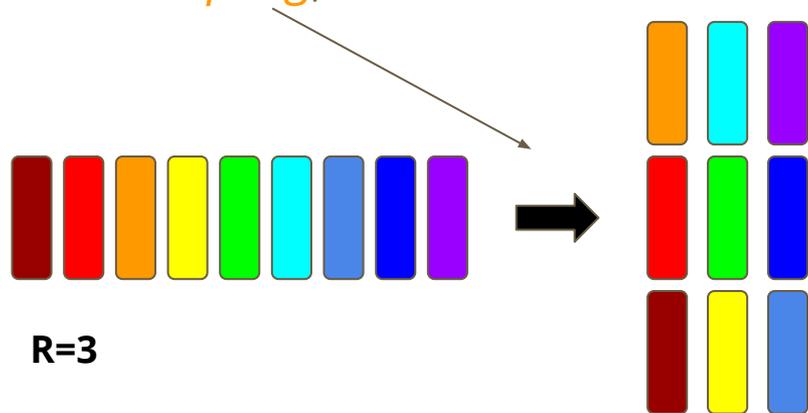
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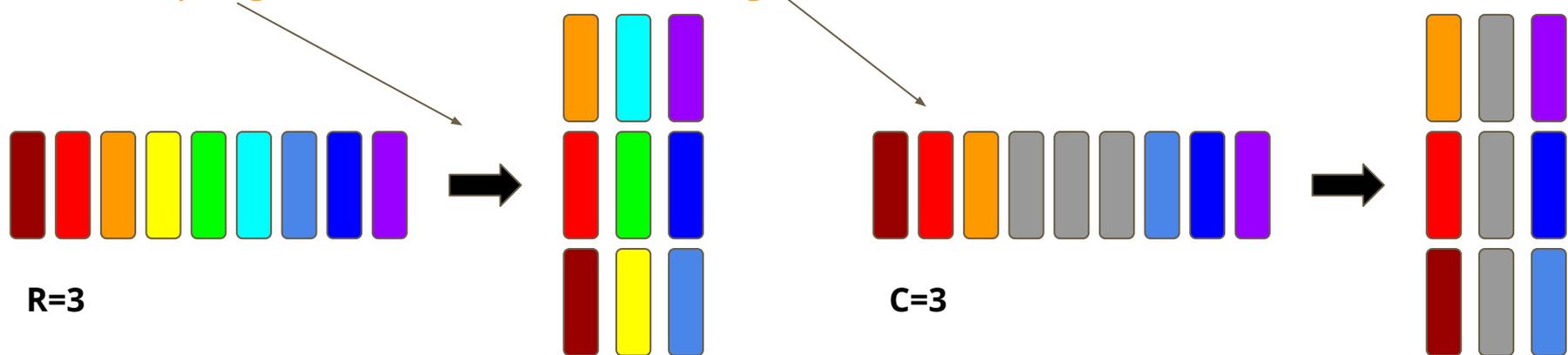
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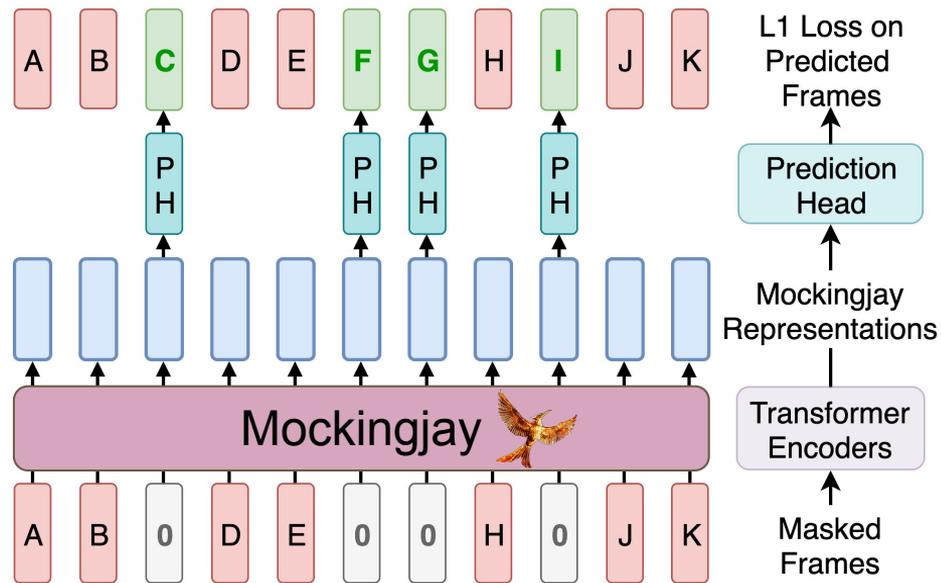


Address the long and smooth problem with:

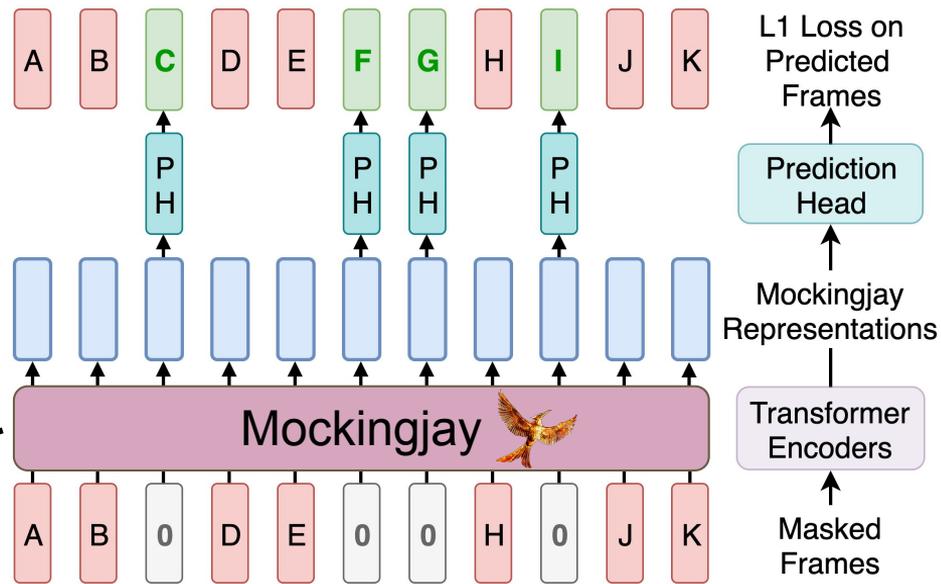
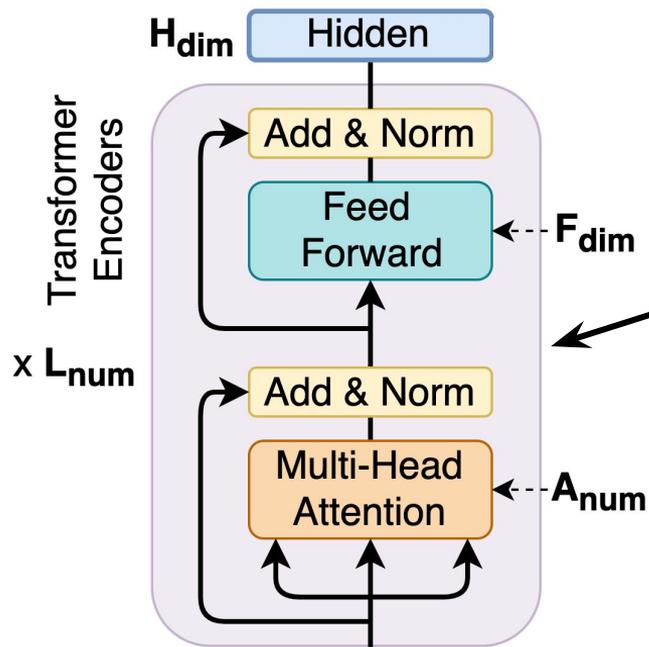
*Downsampling*, and *consecutive masking*



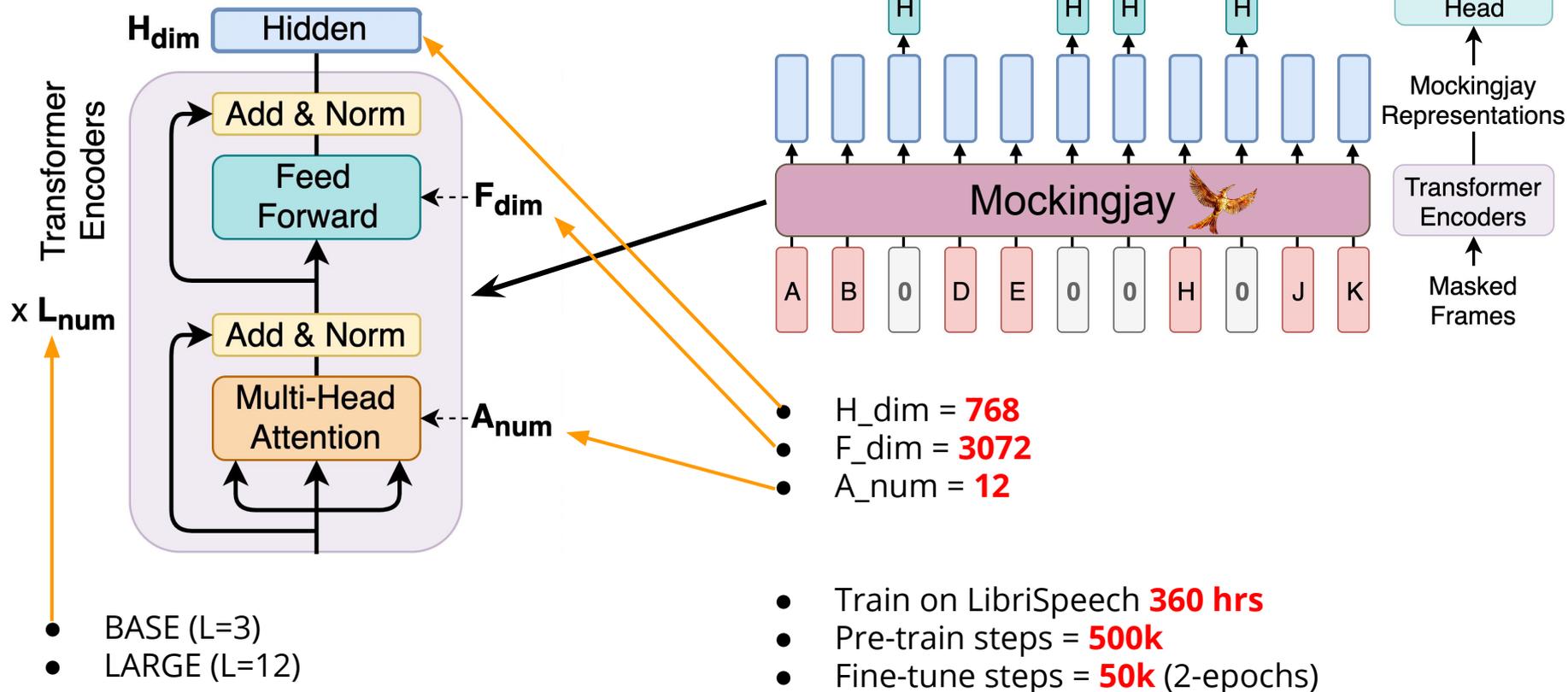
# Model Architecture



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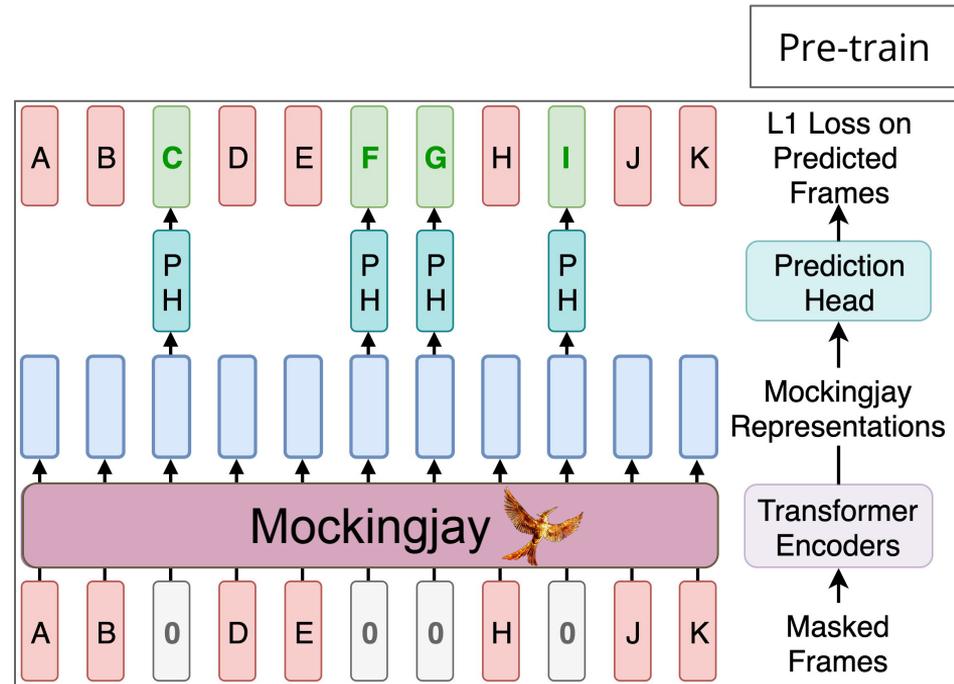


# Incorporating with Downstream Tasks

## 1) Feature Extraction

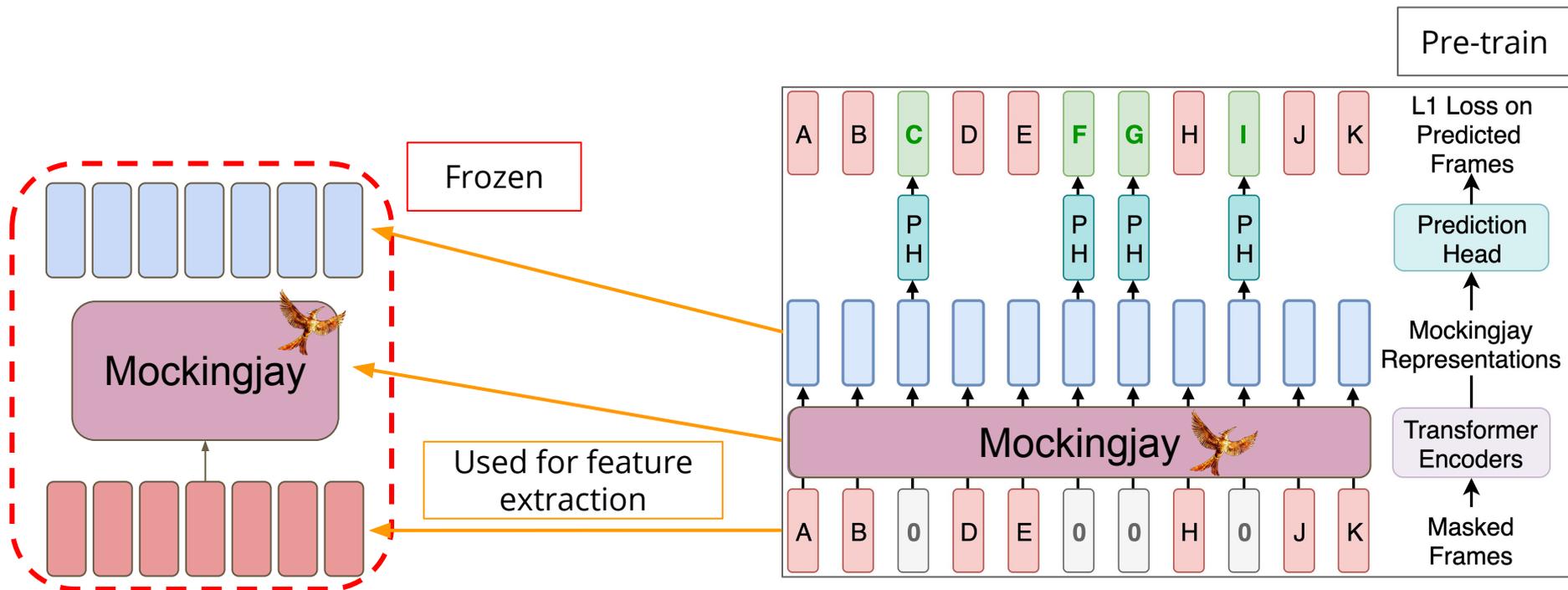
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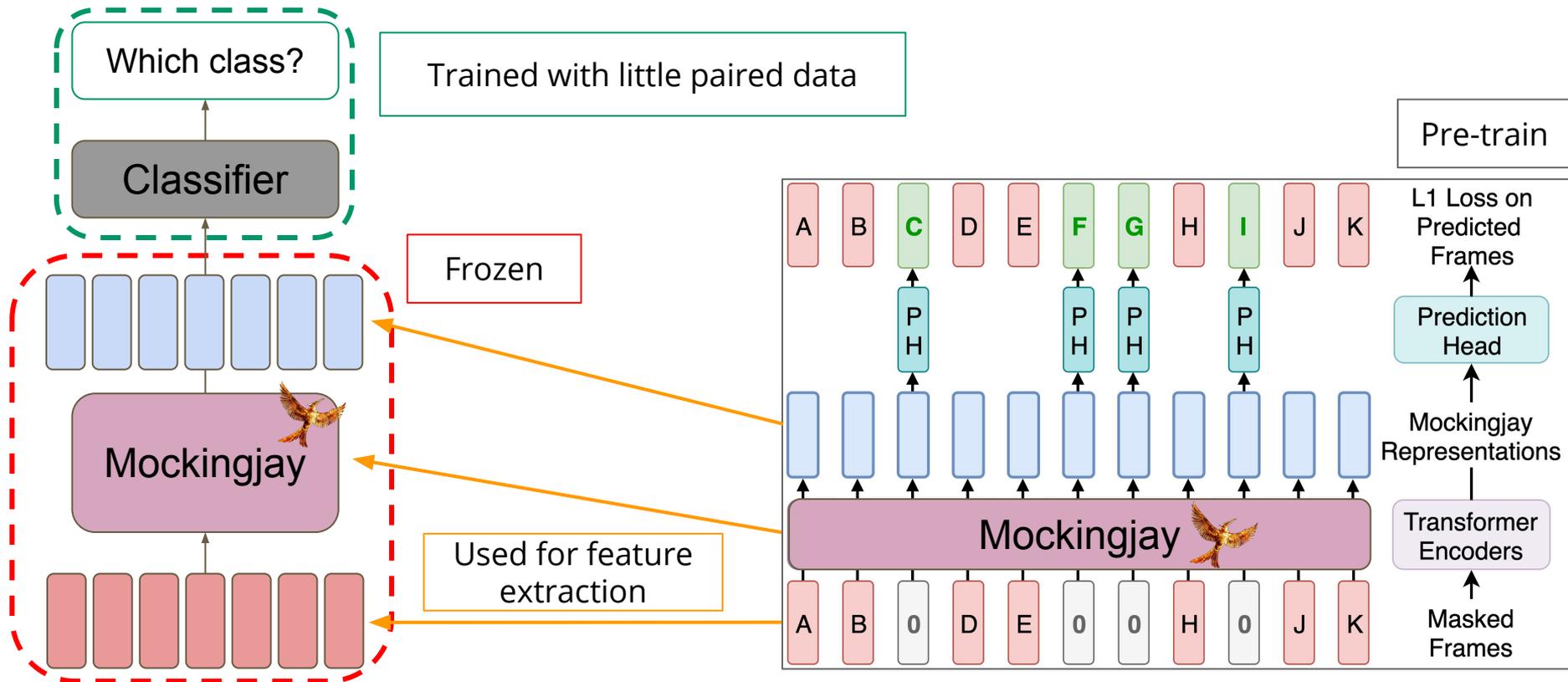
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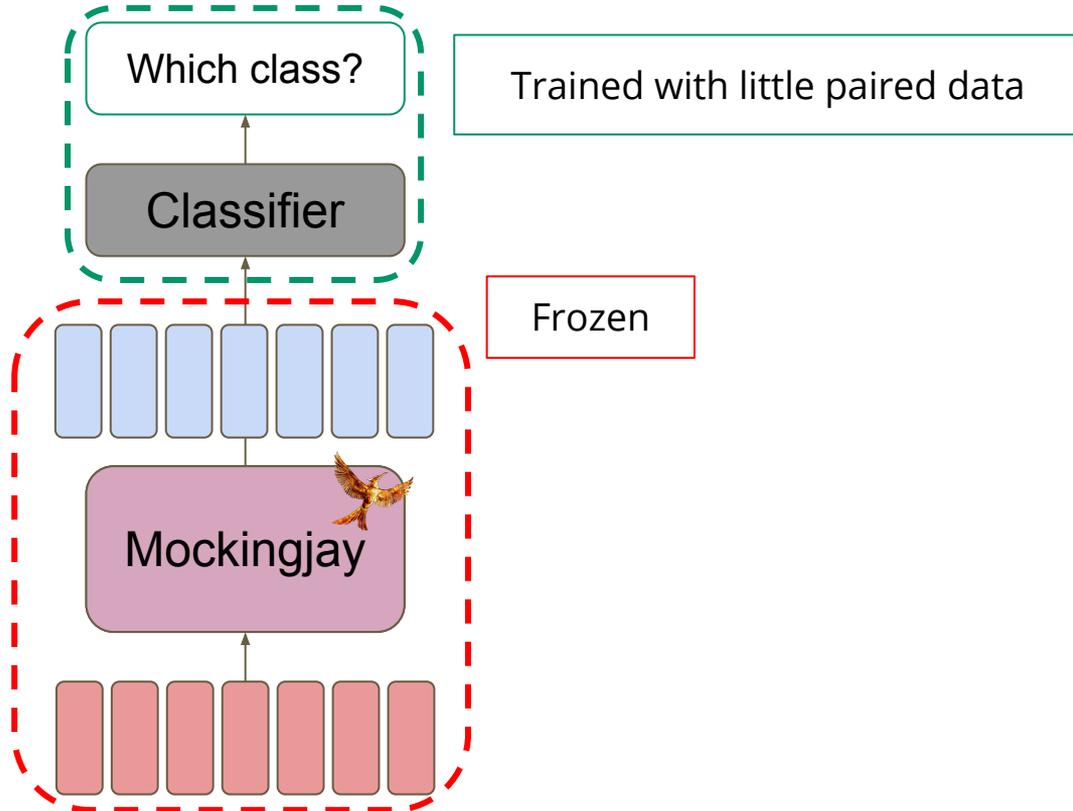
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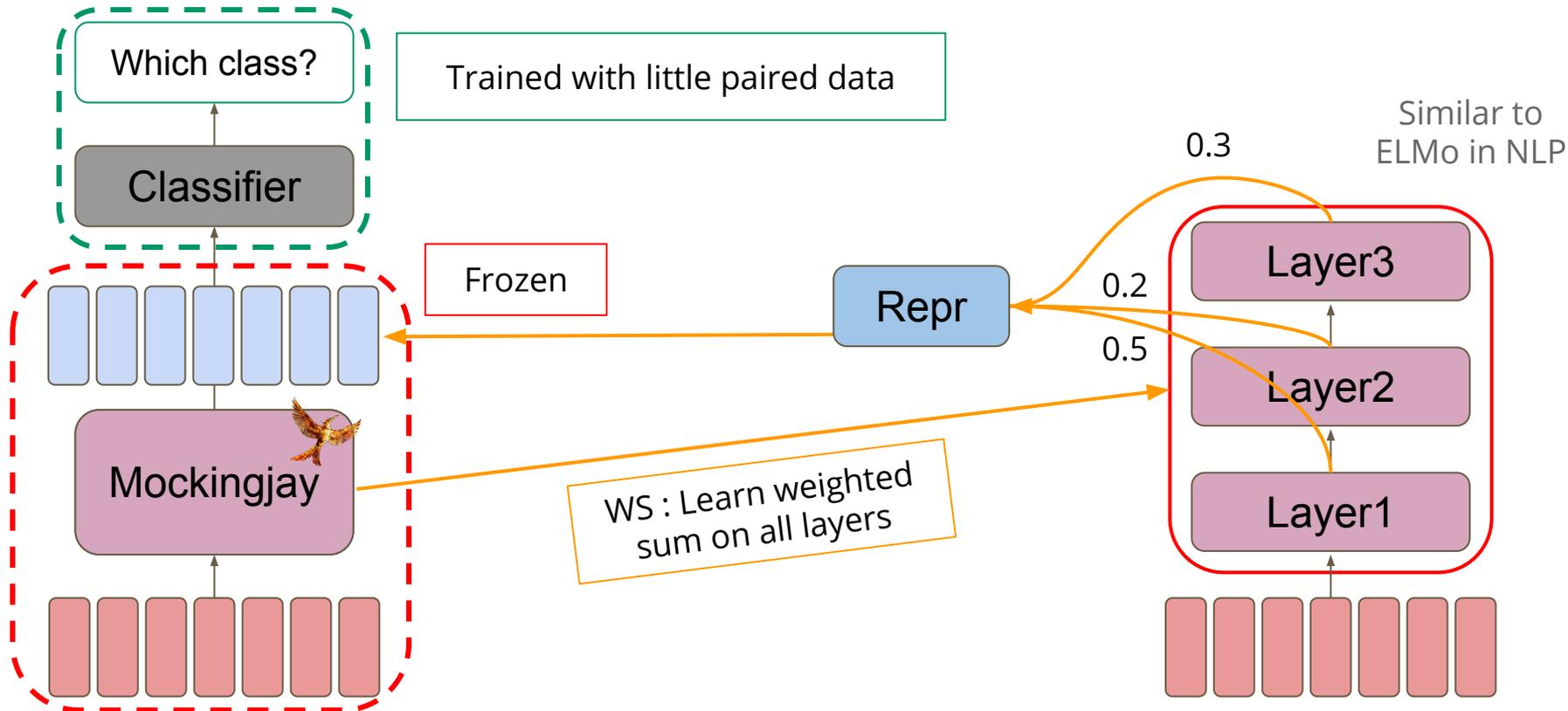
# Incorporating with Downstream Tasks

## 2) Weighted Sum from All Layers (WS)



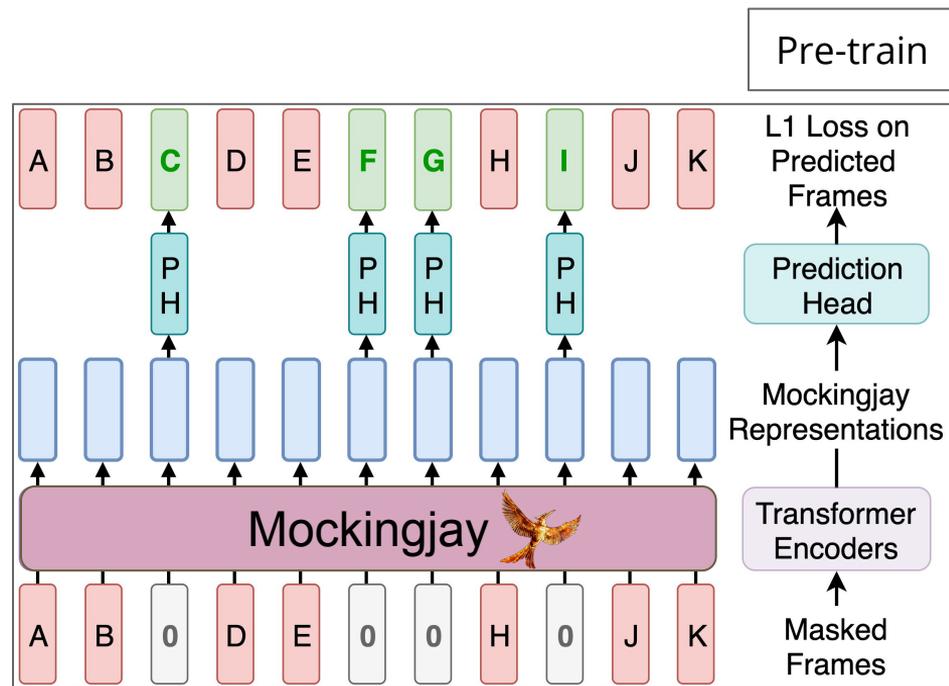
# Incorporating with Downstream Tasks

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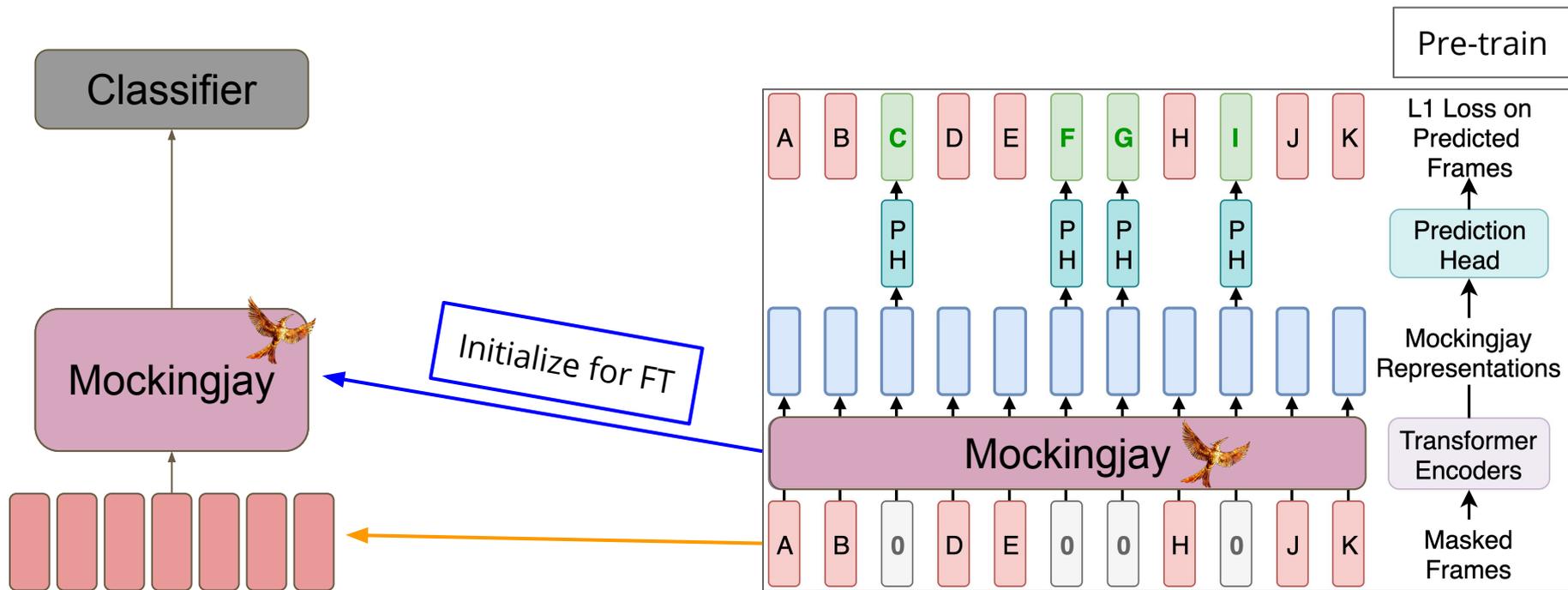
# Incorporating with Downstream Tasks

## 3) Fine-tune (FT2)



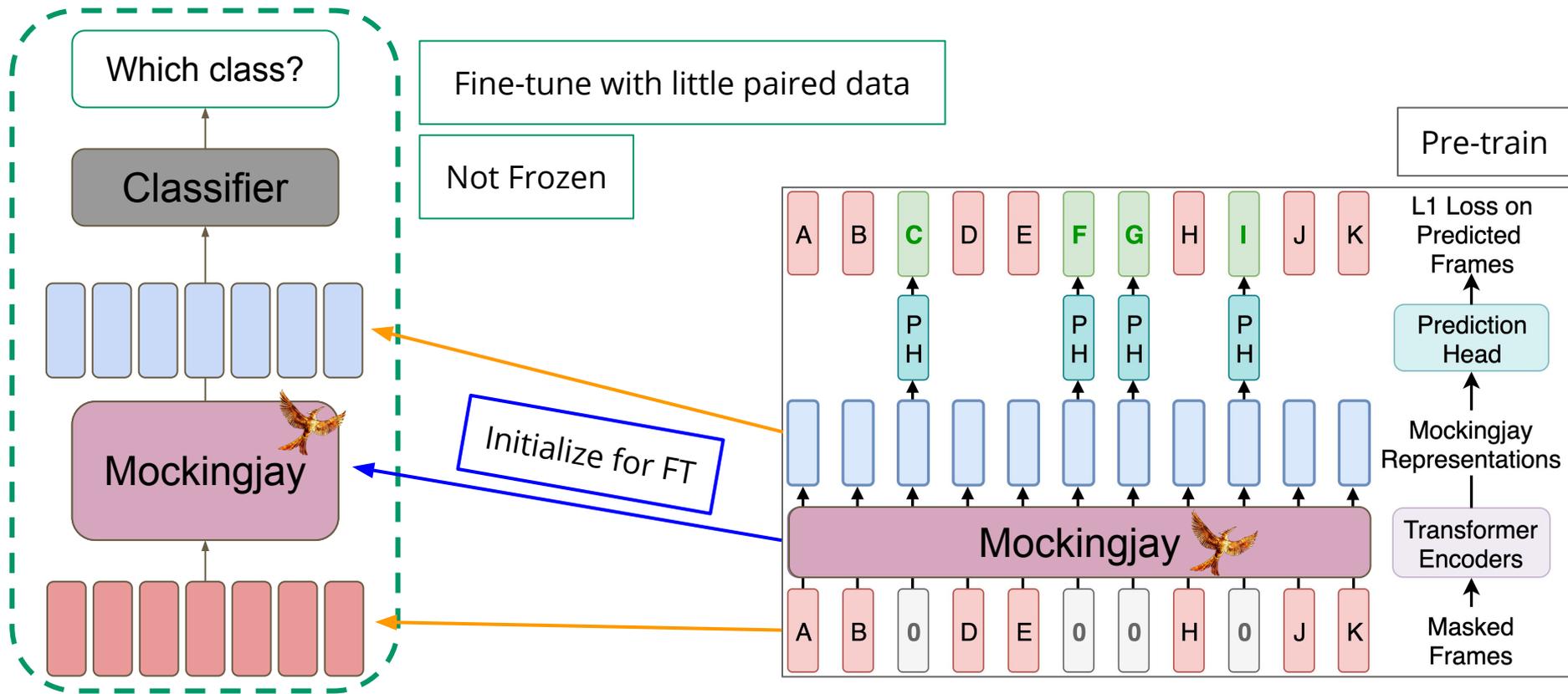
# Incorporating with Downstream Tasks

## 3) Fine-tune (FT2)



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# Experiments

We report results on 3 different downstream tasks:

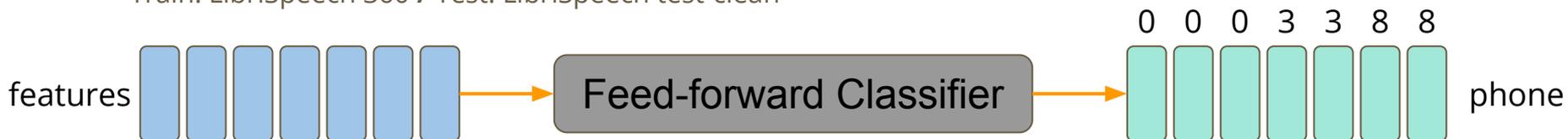
- Phoneme Classification
- Speaker Recognition
- Sentiment Classification on spoken content

# Experiments

We report results on 3 different downstream tasks:

- Phoneme Classification (72 classes):

Train: LibriSpeech 360 / Test: LibriSpeech test-clean



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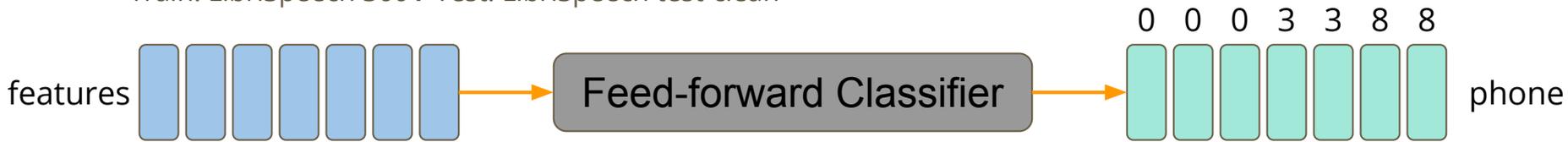
# Experiments

[3] Multimodal language analysis in the wild: CMU-MOSEI dataset and interpretable dynamic fusion graph

We report results on 3 different downstream tasks:

- Phoneme Classification (72 classes):

Train: LibriSpeech 360 / Test: LibriSpeech test-clean



- Speaker Recognition (63 classes):

Train: 90% of LibriSpeech 100 / Test: 10% of LibriSpeech 100

- Sentiment Classification on spoken content (2 classes):

To demonstrate domain invariant transferability, we use another dataset: MOSEI [3]



# Experiments - 1/3

Acoustic Features	Phoneme Classification	Speaker Recognition	Sentiment Classification
Mel Features	49.1	70.1	64.6
BASE	60.9	94.5	67.4
LARGE	<b>64.3</b>	<b>96.3</b>	<b>70.1</b>

Consistent results over all three tasks:  
Mel < BASE < LARGE

# Experiments - 2/3

Acoustic Features	Phoneme Classification	Speaker Recognition	Sentiment Classification
Mel Features	49.1	70.1	64.6
BASE	60.9	94.5	67.4
LARGE	64.3	96.3	70.1
LARGE-WS	<b>69.9</b>	<b>96.4</b>	<b>71.1</b>

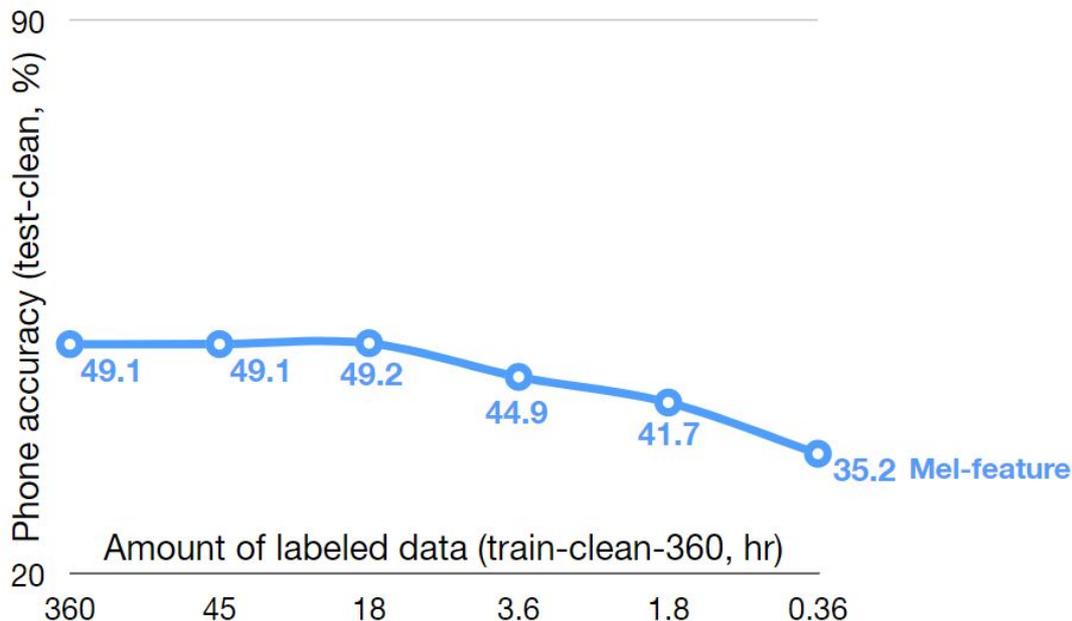
Consistent results over all three tasks:  
LARGE < LARGE-WS

# Experiments - 3/3

Acoustic Features	Phoneme Classification	Speaker Recognition	Sentiment Classification
Mel Features	49.1	70.1	64.6
BASE	60.9	94.5	67.4
LARGE	64.3	96.3	70.1
LARGE-WS	69.9	96.4	<b>71.1</b>
BASE-FT2	<b>84.3</b>	<b>98.1</b>	68.5
APC [2]	74.1	85.9	66.0

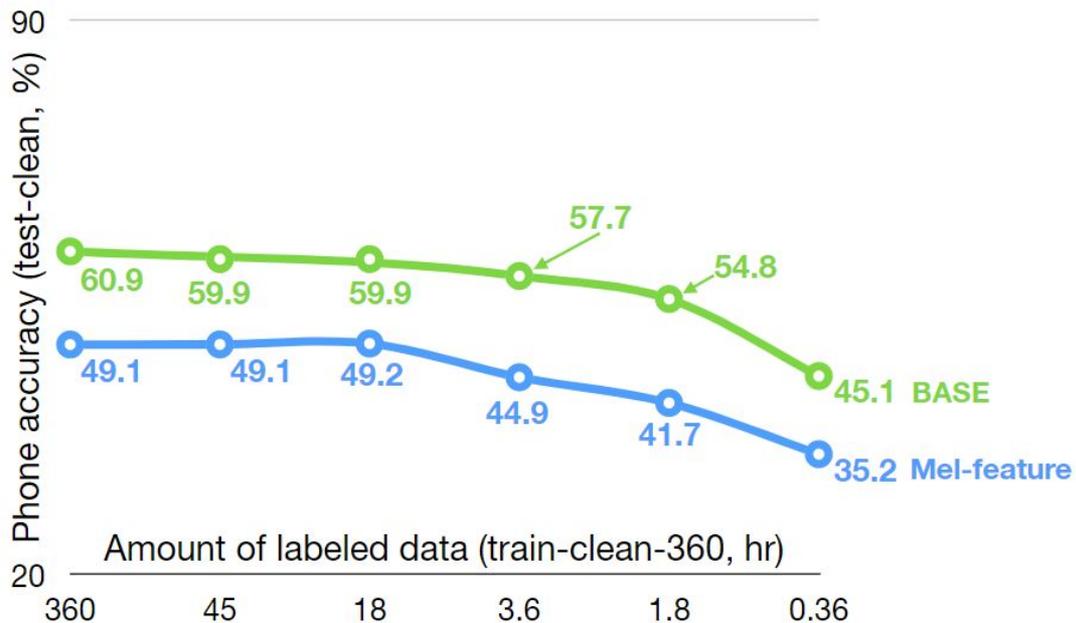
[2] An Unsupervised Autoregressive Model for Speech Representation Learning

# Low-Resource Experiments - 1/6



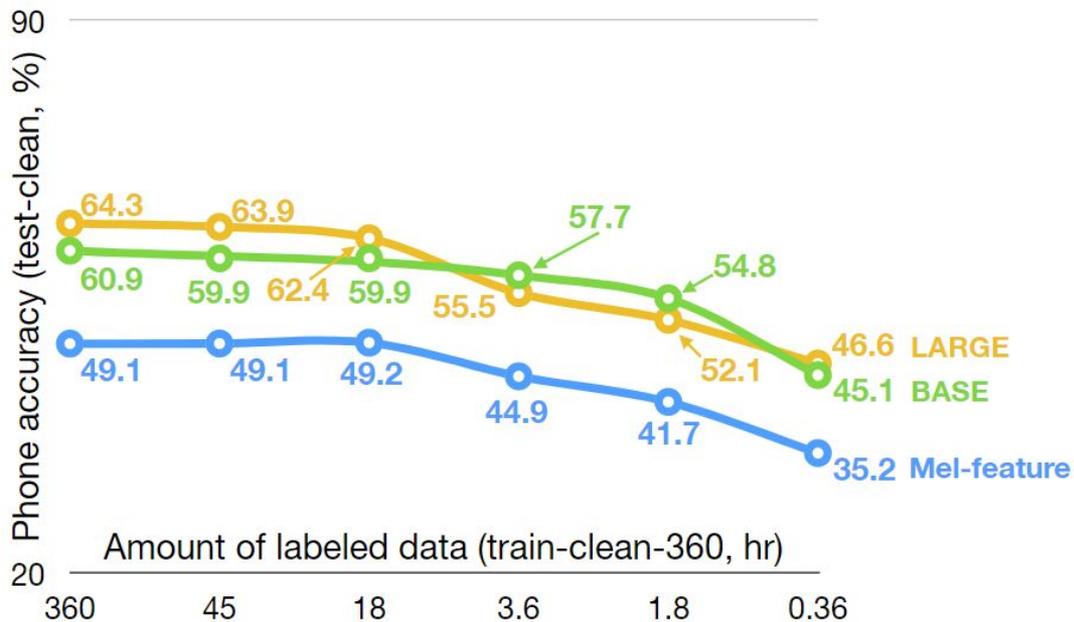
We demonstrate how pre-training on speech can improve supervised training in low resource scenarios, we train with reduced amount of labels.

# Low-Resource Experiments - 2/6



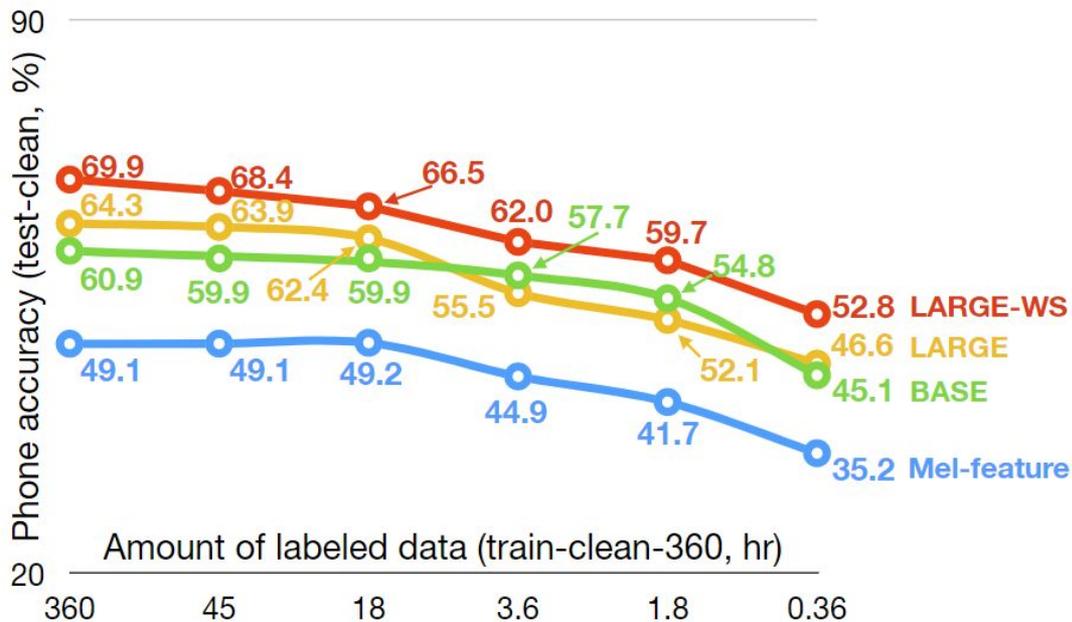
Mel < BASE

# Low-Resource Experiments - 3/6



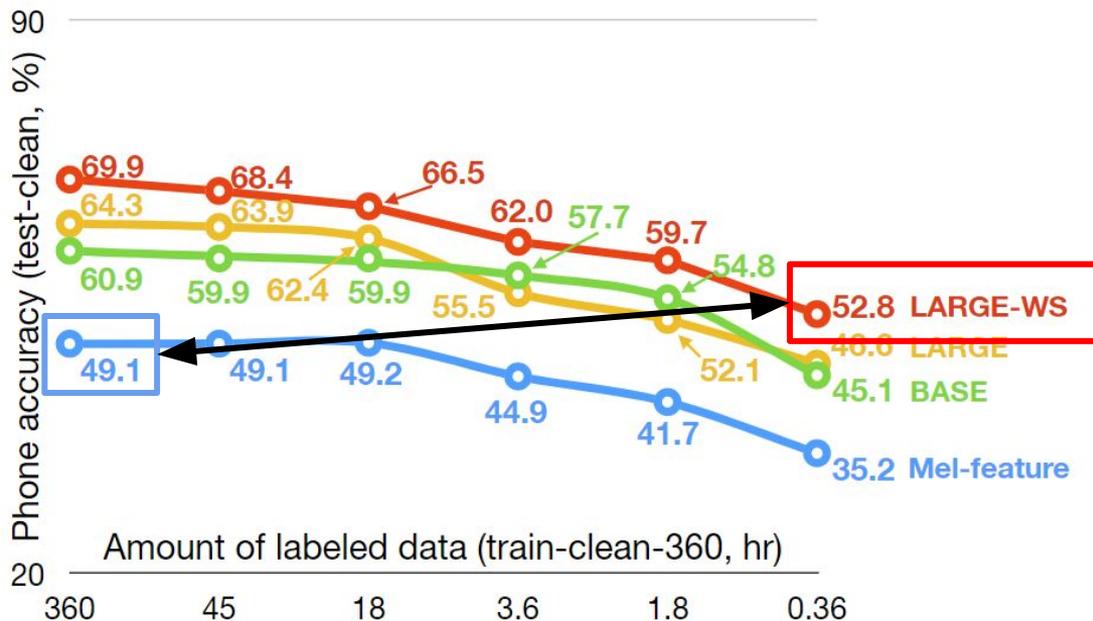
Mel < BASE < LARGE

# Low-Resource Experiments - 4/6



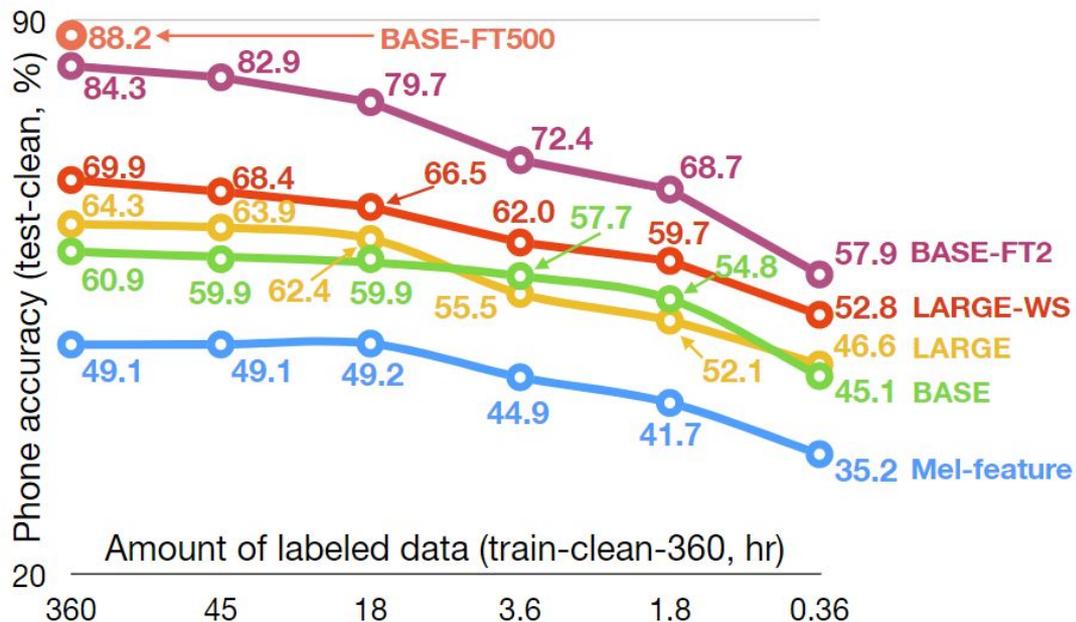
LARGE < LARGE-WS  
with an avg 5.75% improvement

# Low-Resource Experiments - 4/6



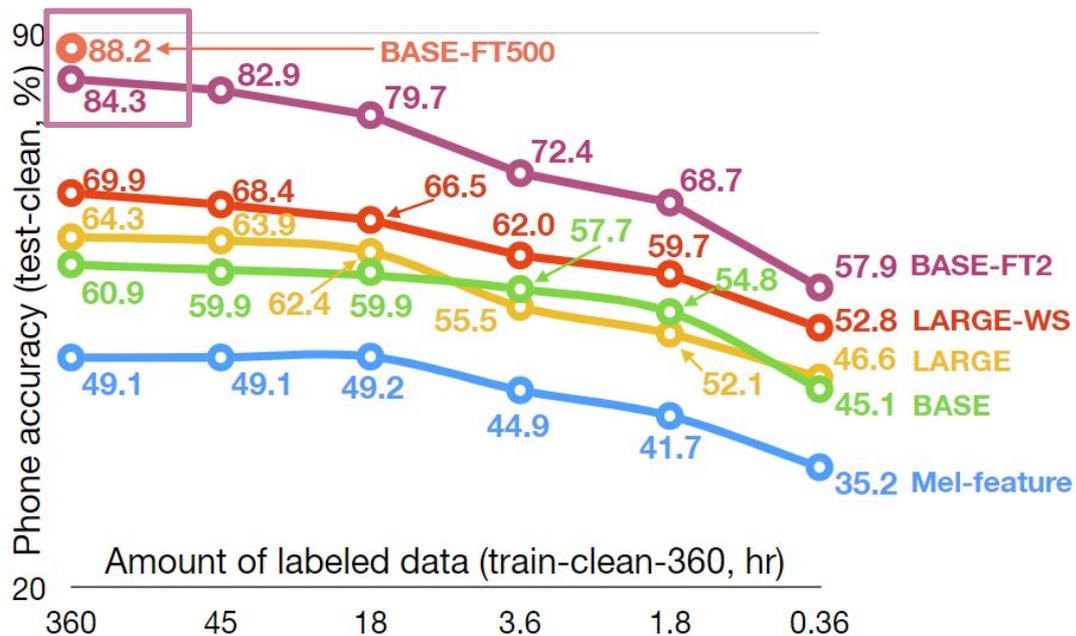
With 0.1% of labels,  
LARGE-WS (52.8%) outperformed Mel (49.1%) that uses all 100% hours of labeled data.

# Low-Resource Experiments - 5/6



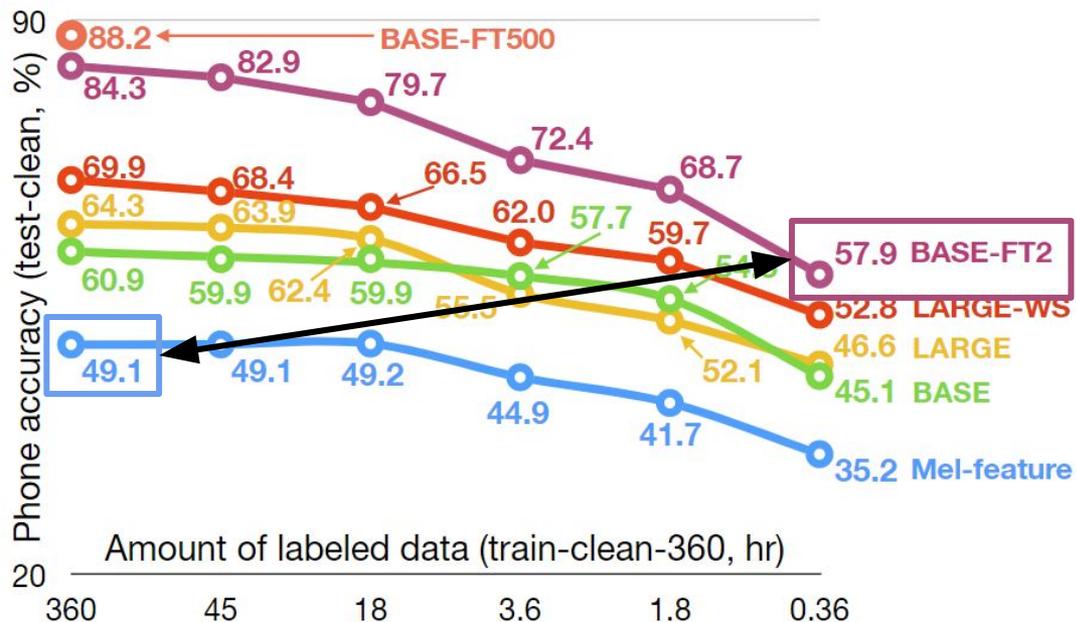
All < BASE-FT2

# Low-Resource Experiments - 5/6



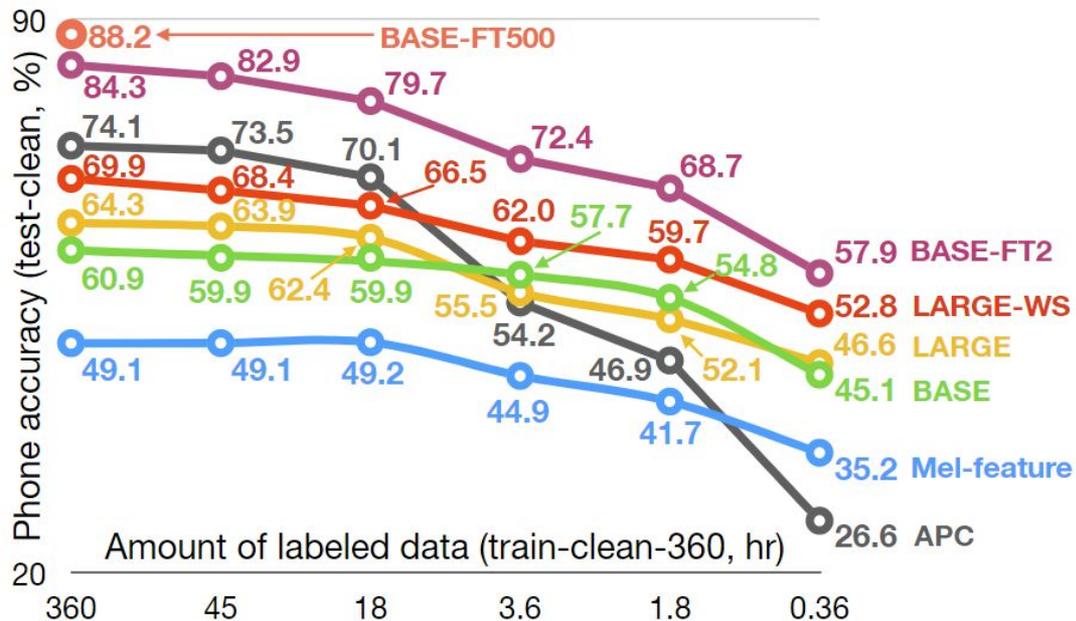
All < BASE-FT2

# Low-Resource Experiments - 5/6



With 0.1% of labels,  
BASE-FT2 (57.9%) outperformed Mel (49.1%) that uses all 100% hours of labeled data.

# Low-Resource Experiments - 6/6



APC works well on full resource but fails to generalize for limited labeled data.

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# Conclusion

— We conclude that unsupervised Mockingjay —  
improves supervised training!

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# Links

This slide (with speaker notes) can be found here:

<https://bit.ly/icassp2020-mockingjay>

Our code and implementation can be found here:

<https://github.com/andi611/Mockingjay-Speech-Representation>