

MATLAB EXPO

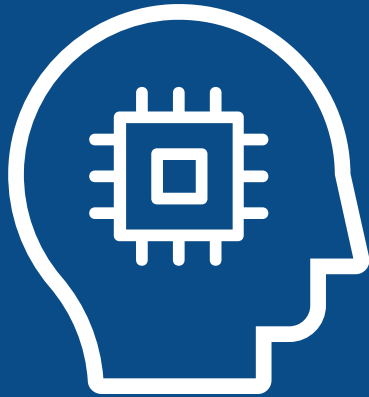
최신 AI 기반 시스템에서 데이터 세트의 중요성
- 음성 인식 AI
장규환, MathWorks



Deep learning is a key technology driving the AI megatrend

ARTIFICIAL INTELLIGENCE

Any technique that enables machines to mimic human intelligence



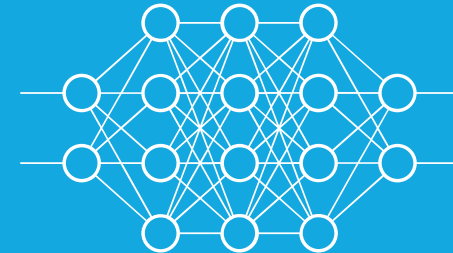
MACHINE LEARNING

Statistical methods that enable machines to “learn” tasks from data without explicitly programming



DEEP LEARNING

Neural networks with many layers that learn representations and tasks “directly” from data



1950s

1980s

2010s

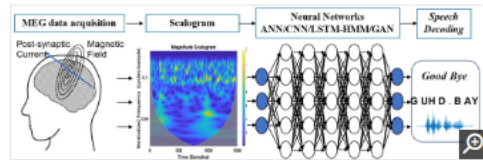
What does it take to develop an effective real-world deep learning system for signal processing applications?

Deep learning use in signal processing applications is growing rapidly

UT Austin Researchers Convert Brain Signals to Words and Phrases Using Wavelets and Deep Learning

"MATLAB is an industry-standard tool, and one that you can trust. It is easier to learn than other languages, and its toolboxes help you get started in new areas because you don't have to start from scratch."

— Dr. Jun Wang, UT Austin



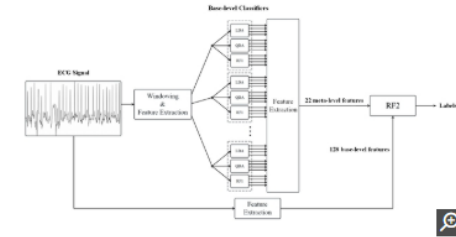
Classifying the brain signals corresponding to the imagined word "goodbye" using feature extraction and deep neural networks.

https://www.mathworks.com/company/user_stories/ut-austin-researchers-convert-brain-signals-to-words-and-phrases-using-wavelets-and-deep-learning.html

MATLAB Based Algorithm Wins the 2017 PhysioNet/CinC Challenge to Automatically Detect Atrial Fibrillation

"I don't think MATLAB has any strong competitors for signal processing and wavelet analysis. When you add in its statistics and machine learning capabilities, it's easy to see why nonprogrammers enjoy using MATLAB, particularly for projects that require combining all these methods."

— Ali Bahrami Rad, Aalto University



Block diagram for Black Swan's atrial fibrillation detection algorithm.

https://www.mathworks.com/company/user_stories/matlab-based-algorithm-wins-the-2017-physionet-cinc-challenge-to-automatically-detect-atrial-fibrillation.html

Shell performs Seismic Event Detection with Deep Learning

Challenges

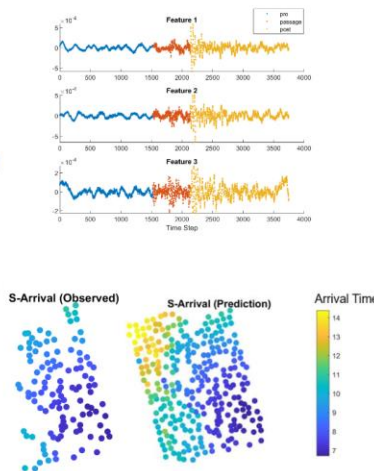
- Terabytes of passive seismic data from geophones
- Traditional methods time/labor intensive (5 months &~ \$100K)
- Event detection inconsistent/unreliable in 'low' signal to noise records

Solution

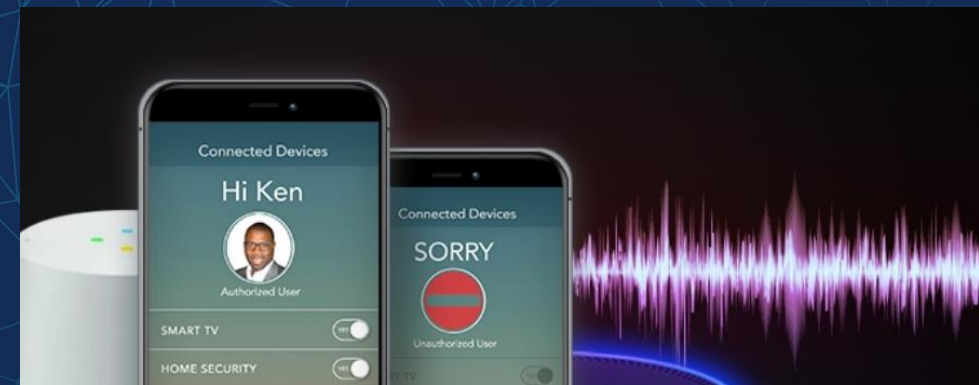
- Train LSTM network to detect P-wave and S-wave arrivals via sequence-to-sequence classification

Results

- >98% accuracy for arrival prediction
- Networks generalizes to other data (sites, source mechanisms)



<https://library.seg.org/doi/pdf/10.1190/segam2019-3215081.1>



Voice Interface: The Touchscreen of the Next Century

How AI and Signal Processing Came Together to Track the DNA of Sound

<https://www.mathworks.com/company/mathworks-stories/ai-signal-processing-for-voice-assistants.html>

A Practical Example: Trigger Word Detection

(The embedded gateway to your cloud-based voice assistant)





44.1.00	44.3.00	45.1.00	45.3.00	46.1.00	46.3.00	47.1.00	47.3.00	48.1.00
1:26.000	1:27.000	1:28.000	1:29.000	1:30.000	1:31.000	1:32.000	1:33.000	1:34.000

1

ROUTE

M S FX

IN Left

VST: Trigger Word Detector (MathWorks) - Track 1

No preset

Param 2 in 2 out UI

Gain 0.000 dB

Chime Level -12.000 dB

Select time

GLOBAL none

FX ROUTING MONO

FX ROUTING

MASTER

Mixer

1

12 -inf -inf

6 -6 -6

6 -18 -6

12 -12 -12

18 -30 -18

24 -24 -24

30 -42 -30

36 -36 -36

42 -54 -42

-inf -inf

Rate: 1.0

Selection: 1.1.00 1.1.00 0.0.00

Find most of the code for this example online

The screenshot shows the MathWorks Help Center interface. At the top, there is a navigation bar with the MathWorks logo and links for Products, Solutions, Academia, Support (highlighted), Community, and Events. Below this is a dark blue header with 'Help Center' on the left and a search box on the right. A secondary navigation bar contains links for Documentation (highlighted), Examples, Functions, Blocks, Apps, Videos, and Answers. The main content area is titled 'Keyword Spotting in Noise Using MFCC and LSTM Networks'. It includes an introduction paragraph, an 'Introduction' section, an 'Example Summary' section, and a numbered list of four steps. A left sidebar contains a 'CONTENTS' menu with a tree view of the page structure, including 'ON THIS PAGE' and various sub-sections like 'Introduction', 'Example Summary', and 'Train the LSTM Network'.

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Help Center Search Support

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- « Documentation Home
- « Audio Toolbox
- « Machine Learning and Deep Learning for Audio
- Keyword Spotting in Noise Using MFCC and LSTM Networks**
- ON THIS PAGE
 - Introduction
 - Example Summary
 - Inspect the Validation Signal
 - Inspect the KWS Baseline
 - Load Speech Commands Data Set
 - Create Training Sentences and Labels
 - Extract Features
 - Extract Features from Training Dataset
 - Extract Validation Features
 - Define the LSTM Network Architecture
 - Define Training Options
 - Train the LSTM Network

Documentation Examples Functions Blocks Apps Videos Answers

Keyword Spotting in Noise Using MFCC and LSTM Networks

This example shows how to identify a keyword in noisy speech using a deep learning network. In particular, the example uses a Bidirectional Long Short-Term Memory (BiLSTM) network and Mel-Frequency Cepstral Coefficients (MFCC).

Introduction

Keyword spotting (KWS) is an essential component of voice-assist technologies, where the user speaks a predefined keyword to wake-up a system before speaking a complete command or phrase.

This example trains a KWS deep network with feature sequences of mel-frequency cepstral coefficients (MFCC). The example also demonstrates how network accuracy in a noisy environment is improved by using data augmentation.

This example uses long short-term memory (LSTM) networks, which are a type of recurrent neural network (RNN) well-suited to study sequence and time-series data. An LSTM network can look at the time sequence in the forward direction, while a bidirectional LSTM layer (`biLstmLayer`) can look at the time sequence in both forward and backward directions.

The example uses the google Speech Commands Dataset to train the deep learning model. To run the example, you must first download the data set. If you do not want to download the data set, you can load the data set in MATLAB® and typing `load("KWSNet.mat")` at the command line.

Example Summary

The example goes through the following steps:

1. Inspect a "gold standard" keyword spotting baseline on a validation signal.
2. Create training utterances from a noise-free dataset.
3. Train a keyword spotting LSTM network using MFCC sequences extracted from those utterances.
4. Check the network accuracy by comparing the validation baseline to the output of the network when applied to the validation signal.

<https://www.mathworks.com/help/audio/examples/keyword-spotting-in-noise-using-mfcc-and-lstm-networks.html>

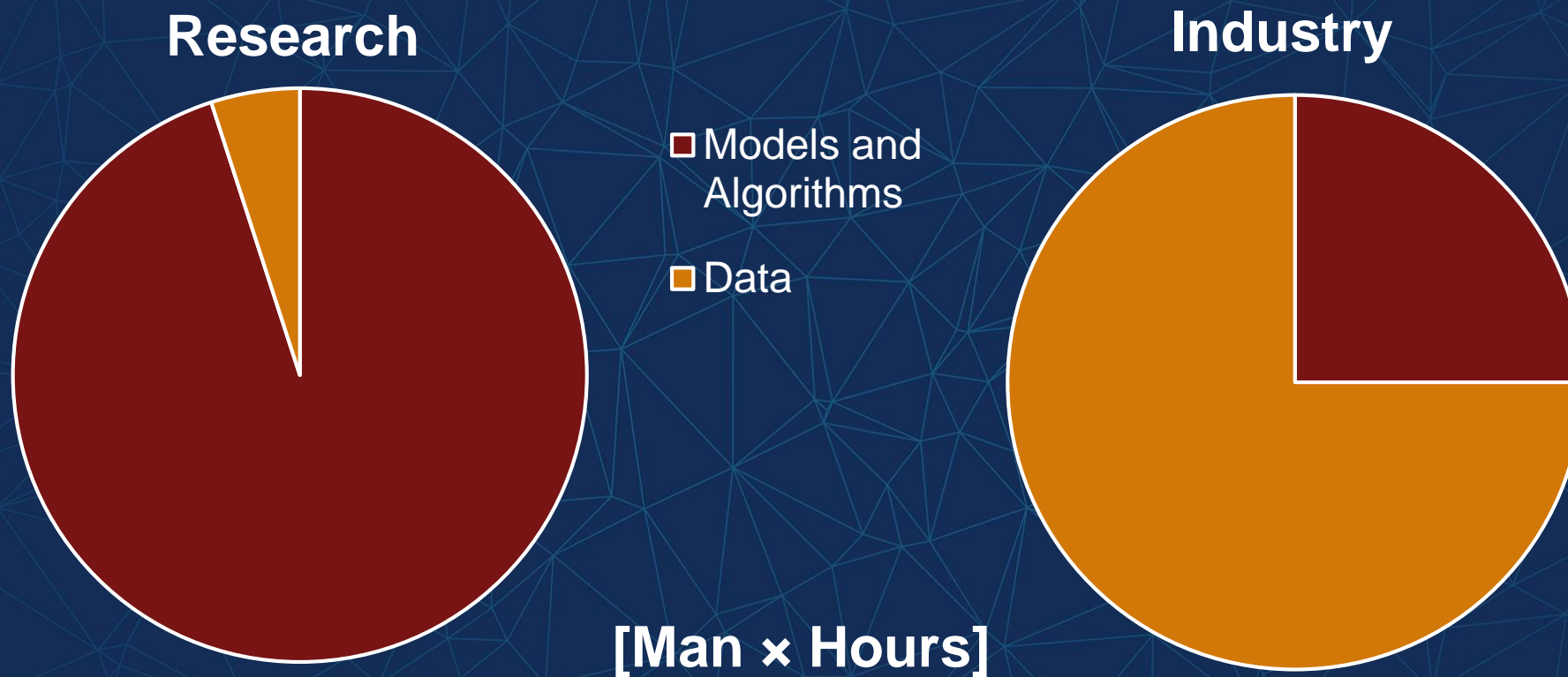
What does it take to develop an effective real-world deep learning system for signal processing applications?

A: "The right deep network design"

"A BiLSTM network with layers of 150 hidden units each, followed by one fully-connected layer and a softmax layer"

A: "A lot of data, a good dose of signal processing expertise, and the right tools for the specific application in hand"

Data Investments in Deep Learning Research vs. Industry



CREATE AND ACCESS DATASETS

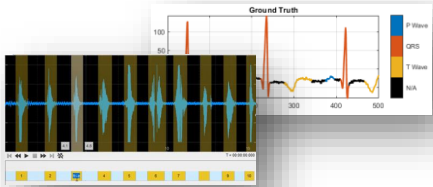
Data sources



Simulation and augmentation

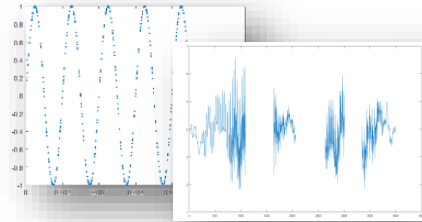


Data Labeling

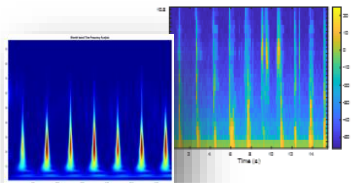


PREPROCESS AND TRANSFORM DATA

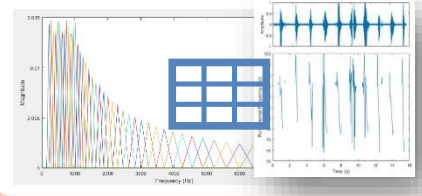
Pre-Processing



Transformation



Feature extraction

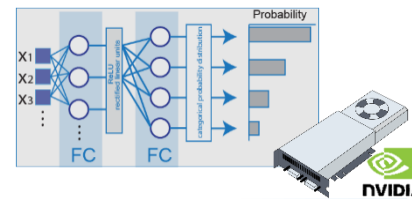


DEVELOP PREDICTIVE MODELS

Import Reference Models/ Design from scratch



Hardware-Accelerated Training

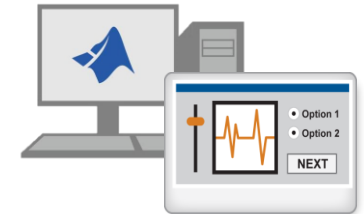


Analyze and tune hyperparameters



ACCELERATE AND DEPLOY

Desktop Apps



Enterprise Scale Systems

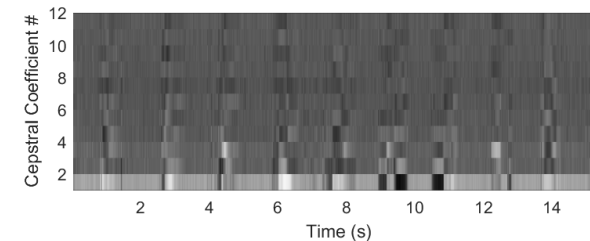
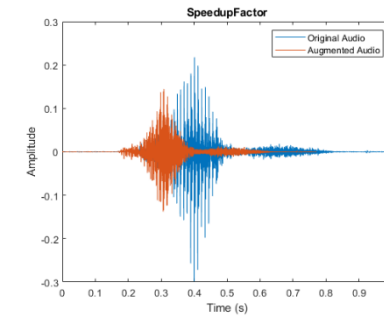
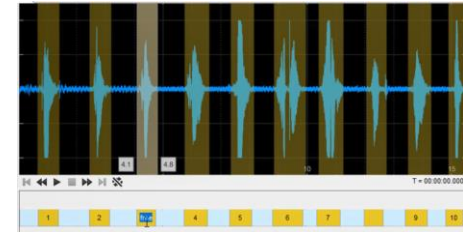
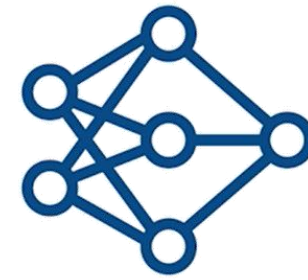
Java
MATLAB
C/C++
Python

Embedded Devices and Hardware



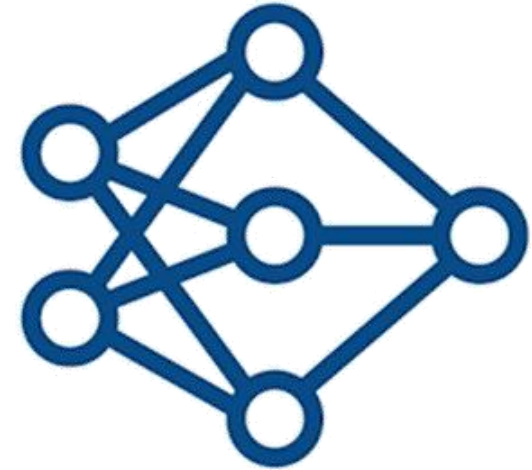
Agenda

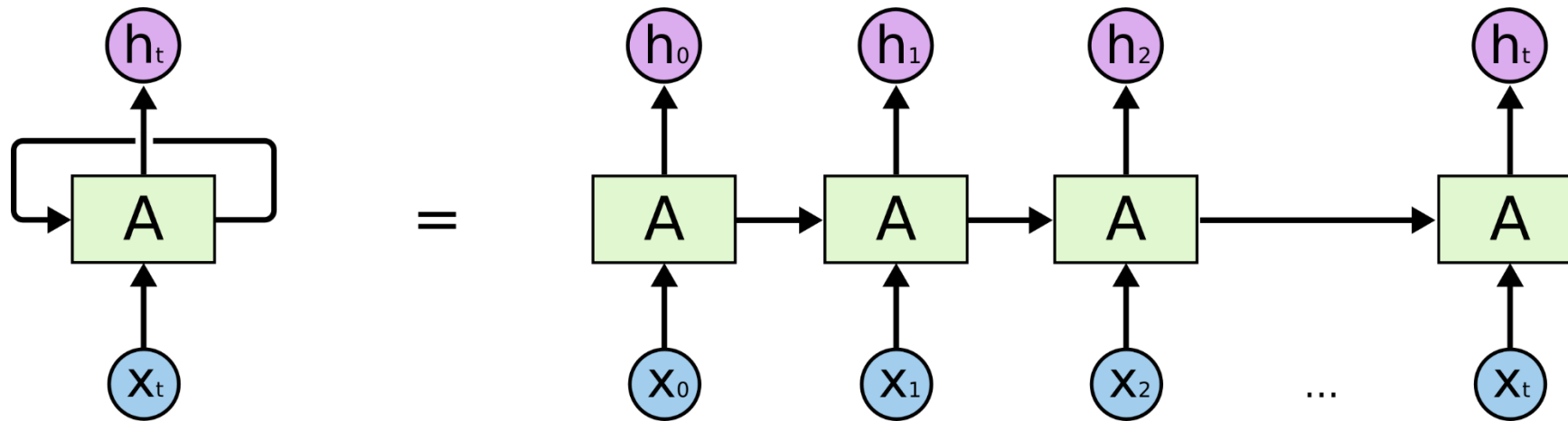
- Basics on training deep neural networks for signals
- Annotating data to train networks for practical applications
- Generating new data – synthesis and augmentation
- Creating inputs for deep networks
- From system models to real-time prototypes



Defining a deep network architecture

```
layers = [ ...  
    sequenceInputLayer(numFeatures)  
    bilstmLayer(150, "OutputMode", "sequence")  
    bilstmLayer(150, "OutputMode", "sequence")  
    fullyConnectedLayer(2)  
    softmaxLayer  
    classificationLayer  
];
```





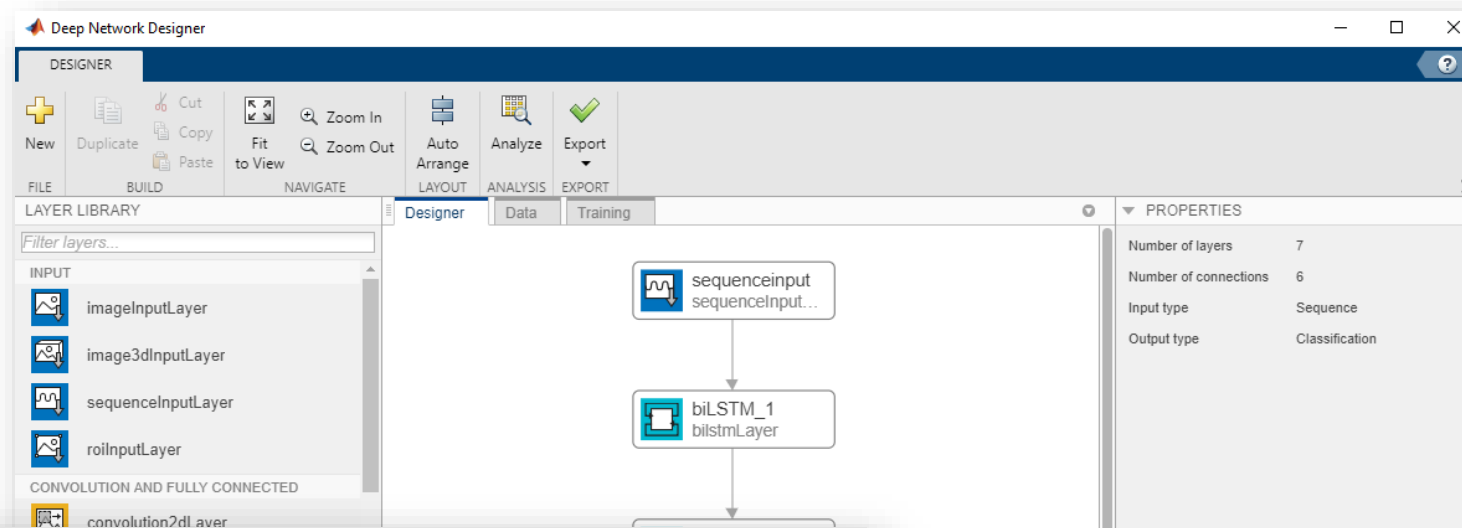
Long Short Term Memory (**LSTM**) Layer

(Recursive Neural Networks, **RNN**)

Defining a deep network architecture

```

layers = [ ...
    sequenceInputLayer(numFeatures)
    biLSTMLayer(150, "OutputMode", "sequence")
    biLSTMLayer(150, "OutputMode", "sequence")
    fullyConnectedLayer(2)
    softmaxLayer
    classificationLayer
];
    
```



ANALYSIS RESULT					
	Name	Type	Activations	Learnables	Total Learnables
1	sequenceinput Sequence input with 42 dimensions	Sequence Input	42	-	0
2	biLSTM_1 BiLSTM with 150 hidden units	BiLSTM	300	InputWeights 1200×42 RecurrentWeights 1200×150 Bias 1200×1	231600
3	biLSTM_2 BiLSTM with 150 hidden units	BiLSTM	300	InputWeights 1200×300 RecurrentWeights 1200×150 Bias 1200×1	541200
4	fc 2 fully connected layer	Fully Connected	2	Weights 2×300 Bias 2×1	602
5	softmax softmax	Softmax	2	-	0
6	classoutput crossentropyex	Classification Output	-	-	0

Start from published recipes...

...or import models developed by others (including from different frameworks)

Long Short-Term Memory Recurrent Neural Network Architectures for Large Scale Acoustic Modeling

Haşim Sak, Andrew Senior, Françoise Beaufays

Google, USA

Long short-term memory for speaker generalization in supervised speech separation

Jitong Chen^{a)} and DeLiang Wang^{b)}

Department of Computer Science and Engineering, The Ohio State University, Columbus, Ohio 43210, USA

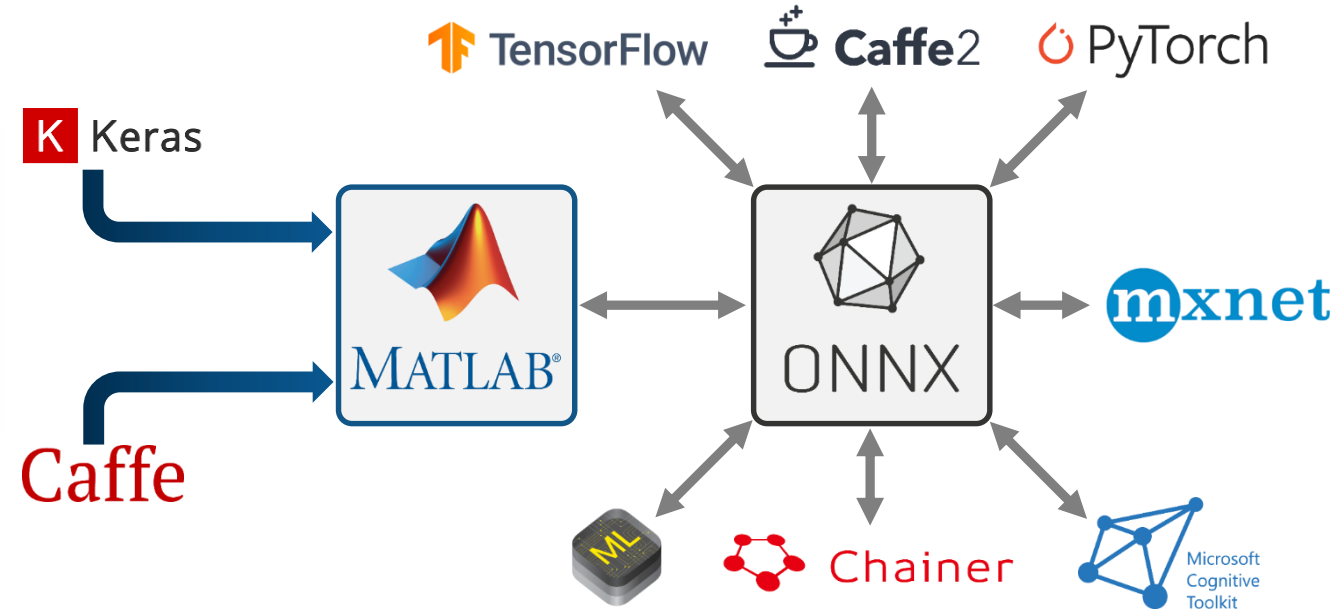
An Improved Residual LSTM Architecture for Acoustic Modeling

Lu Huang
Department of Electronic Engineering
Tsinghua University
Beijing, China
e-mail: huanglu.th@gmail.com

Jiasong Sun
Department of Electronic Engineering
Tsinghua University
Beijing, China
e-mail: sunjiasong@tsinghua.edu.cn

Ji Xu
Department of Speech Acoustics & Content Understanding
Institute of Acoustics, Chinese Academy of Sciences

Yi Yang
Department of Electronic Engineering
Tsinghua University



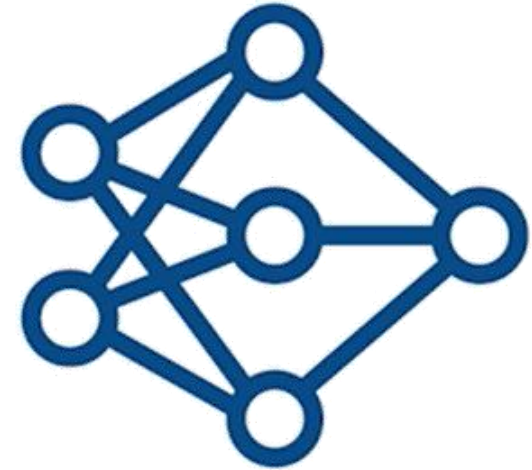
* Random examples found via web search
(No endorsement implied)

Training a deep network

```
layers = [ ...
    sequenceInputLayer(numFeatures)
    bilstmLayer(150, "OutputMode", "sequence")
    bilstmLayer(150, "OutputMode", "sequence")
    fullyConnectedLayer(2)
    softmaxLayer
    classificationLayer
];

maxEpochs      = 10;
miniBatchSize  = 64;
options = trainingOptions("adam", ...
    "InitialLearnRate", 1e-4, ...
    "MaxEpochs", maxEpochs, ...
    "MiniBatchSize", miniBatchSize, ...
    "Shuffle", "every-epoch", ...
    "Verbose", false, ...
    "ValidationFrequency", floor(numel(TrainingFeatures)/miniBatchSize), ...
    "ValidationData", {FeaturesValidationClean.', BaselineV}, ...
    "Plots", "training-progress", ...
    "LearnRateSchedule", "piecewise", ...
    "LearnRateDropFactor", 0.1, ...
    "LearnRateDropPeriod", 5);

[net, info] = trainNetwork(TrainingFeatures, TrainingMasks, layers, options);
```



/ > home > matlab > Documents > AudioWebinar > Code

Workspace

Name	Value
expectedNumPartitions	128
klstm	4
kovlp	4
loadFeatures	1
LSTMSize	150
LSTMSizes	[75,100,125,150]
LSTMSizes	[75,100,125,150]
M	1x4 cell
net	1x1 SeriesNetwork
netLayers	6x1 Layer

Current Folder

Command Window

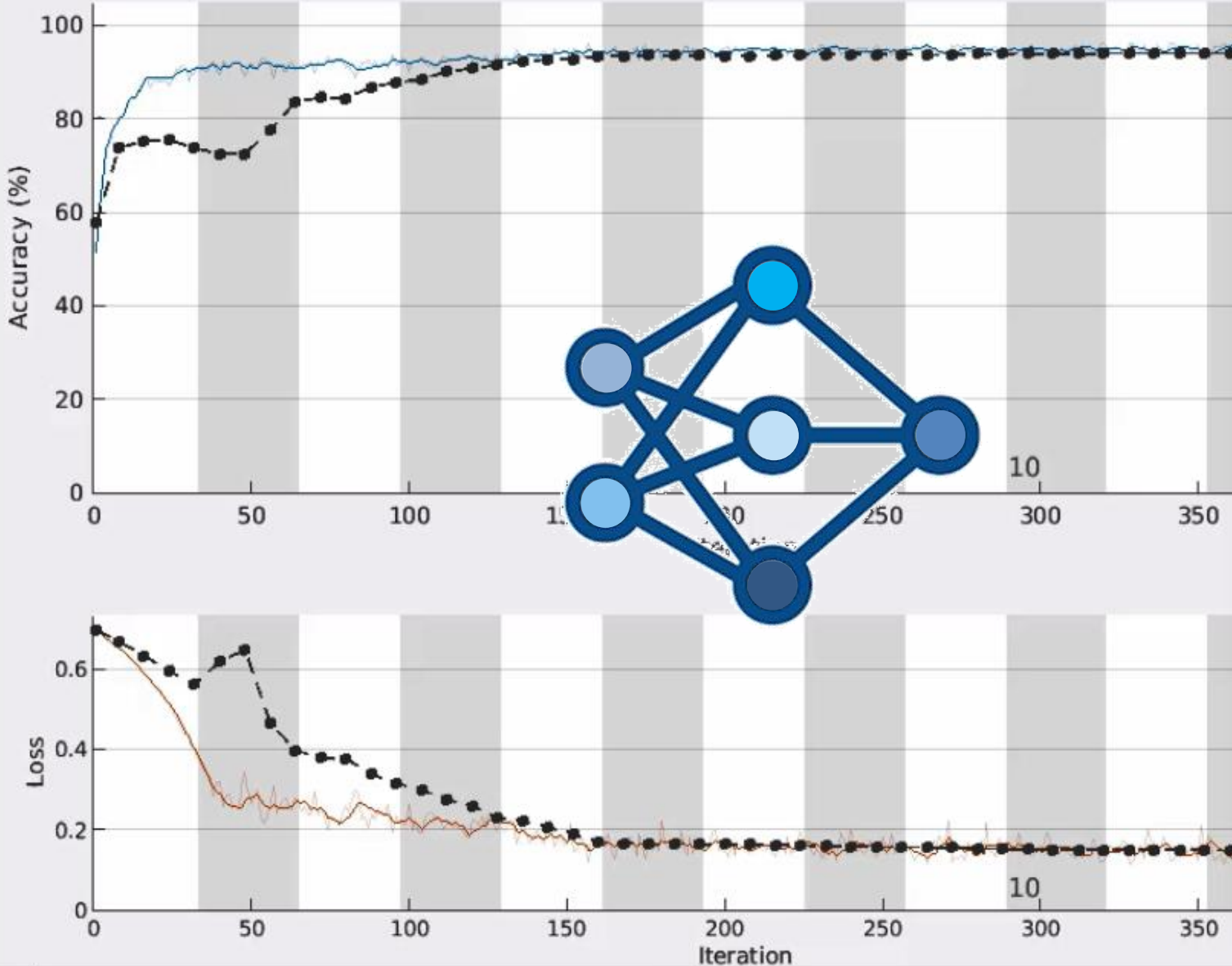
fx >>

Editor - TrainSingleNetwork.m

Variables - feat

```
40 - netLayers = [ ...
41     sequenceInputLayer(numFeatures)
42     bilstmLayer(LSTMSizes(klstm),"OutputMode","sequence")
43     bilstmLayer(LSTMSizes(klstm),"OutputMode","sequence")
44     fullyConnectedLayer(2)
45     softmaxLayer
46     classificationLayer
47 ];
48
49 - trainOptions = trainingOptions("adam", .x.
50     "InitialLearnRate",1e-4, ...
51     "MaxEpochs",12, ...
52     "MiniBatchSize",4, ...
53     "Shuffle","every-epoch", ...
54     "Verbose",false, ...
55     "ValidationFrequency",8, ...
56     "ValidationData",{ValidationFeatures{kovlp},ValidationMasks{kovlp}}, ...
57     "Plots","training-progress", ...
58     "LearnRateSchedule","piecewise", ...
59     "LearnRateDropFactor",0.1, ...
60     "LearnRateDropPeriod",5,...
61     "SequenceLength","Shortest");
62
63 %% Network training
64
65 - tic;
66 - net = trainNetwork(trainingFeatures,trainingMasks,netLayers,trainOptions);
67 - fprintf('Training the network took %g s\n',toc);
68
69
```

Training Progress (20-Mar-2020 12:21:25)



Results

Validation accuracy: 93.96%

Training finished: Reached final iteration

Training Time

Start time: 20-Mar-2020 12:21:25

Elapsed time: 8 min 15 sec

Training Cycle

Epoch: 12 of 12

Iteration: 384 of 384

Iterations per epoch: 32

Maximum iterations: 384

Validation

Frequency: 8 iterations

Patience: Inf

Other Information

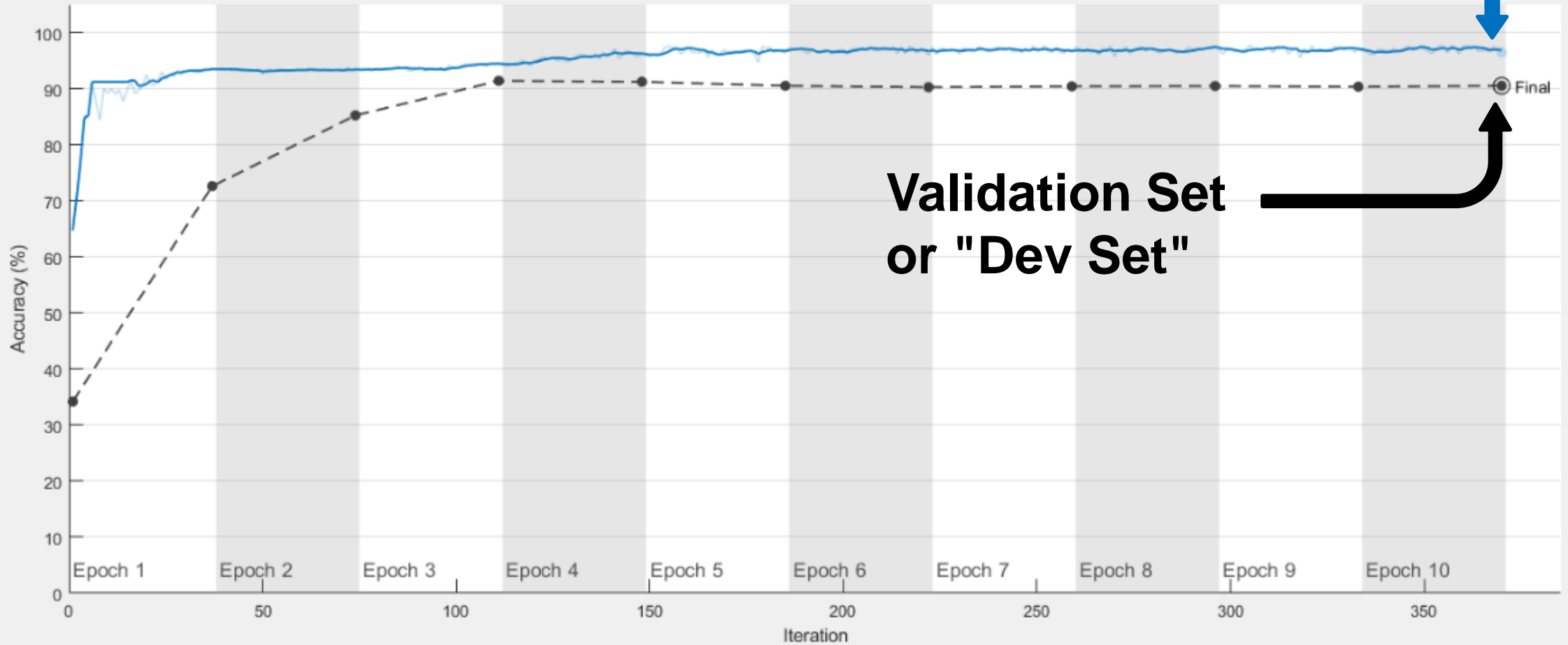
Hardware resource: Single GPU

Learning rate schedule: Piecewise

Accuracy

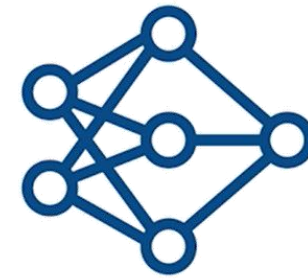
- Training (smoothed)
- Training
- Validation

Training Set



Validation Set
or "Dev Set"

Agenda



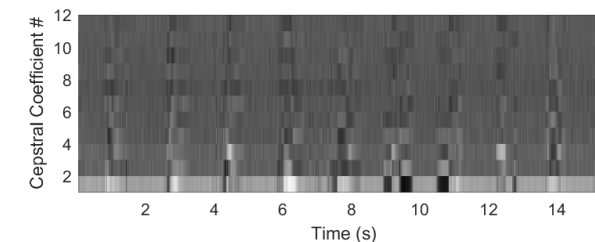
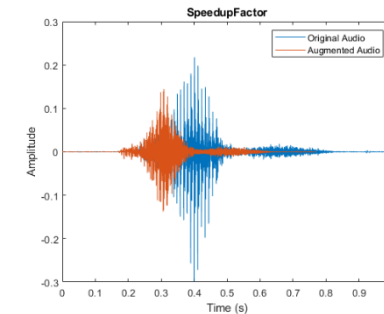
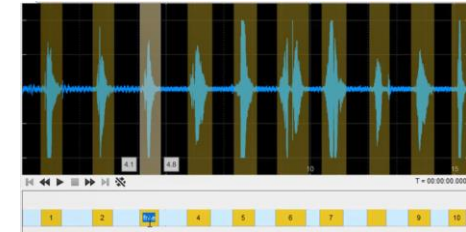
- Basics on training deep neural networks for signals

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Training, Validation, and Test Data

Your full dataset (All of your **data + labels**)

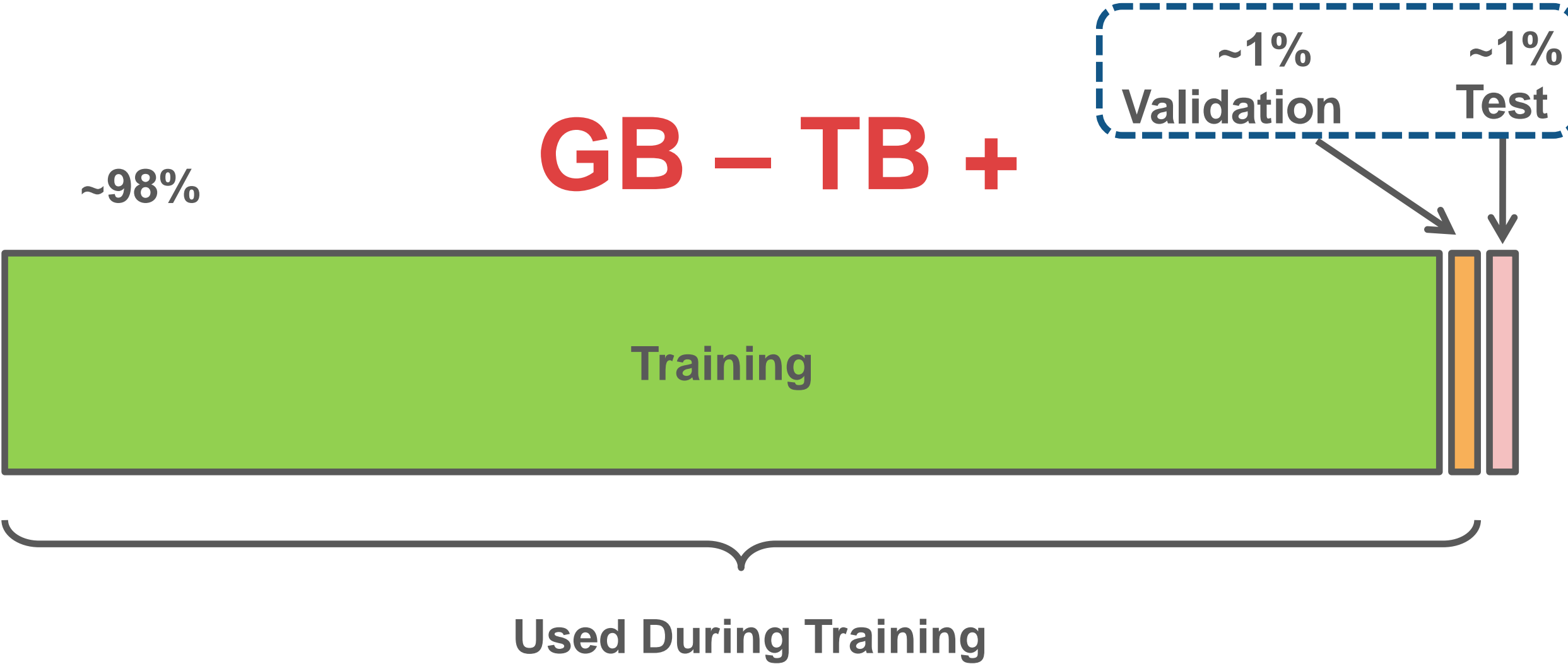
Training, Validation, and Test Data

GB - **TB** +
~60% **KB - MB** ~20% ~20%

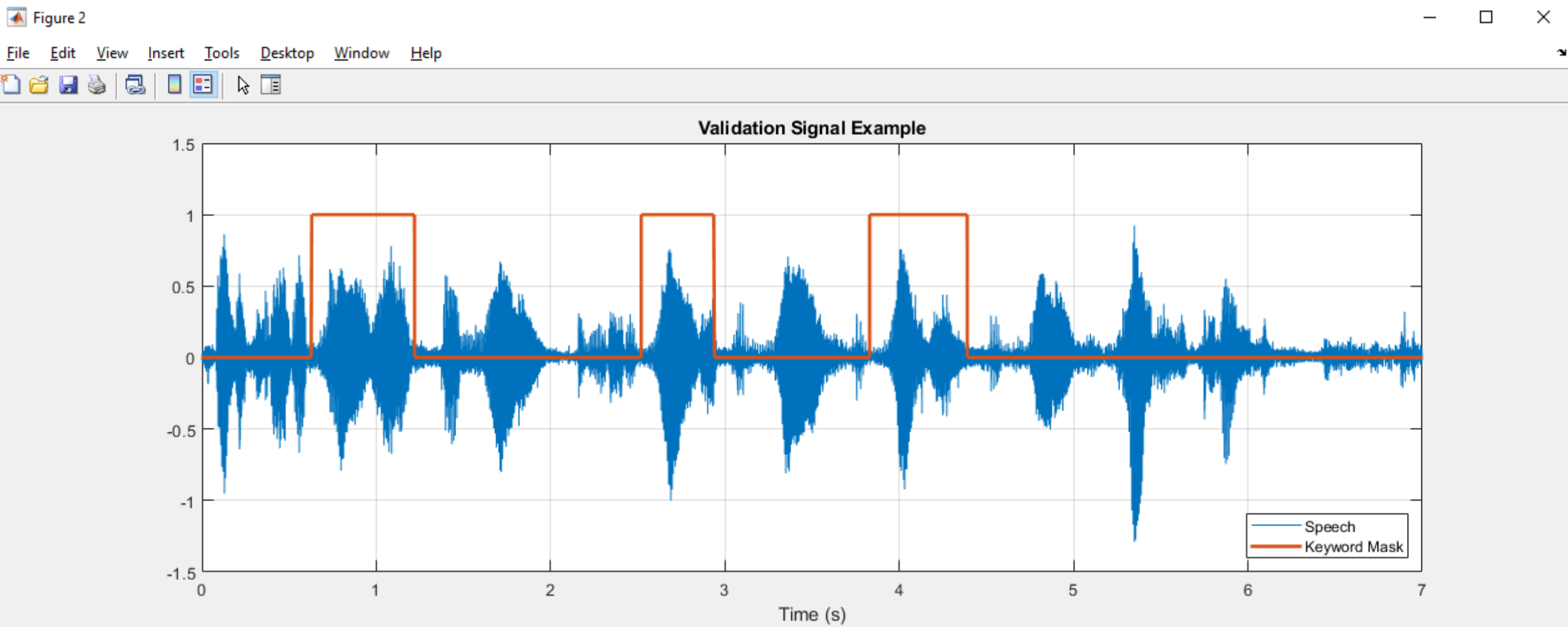


Used During Training

Training, Validation, and Test Data



A good validation data sample – Realistic recording, accurately labeled



How to label new non-annotated data?

Use an intelligent system trained to carry out a similar tasks with proven accuracy!

For example:

- Humans

LABEL RECORD

Load Save Import

Audio Player: Primary Sou...

Settings

Default Layout

Legend

Speech Detector Speech to Text

Export

FILE DEVICE VIEW AUTOMATION EXPORT

Data Browser

▼ Audio Files

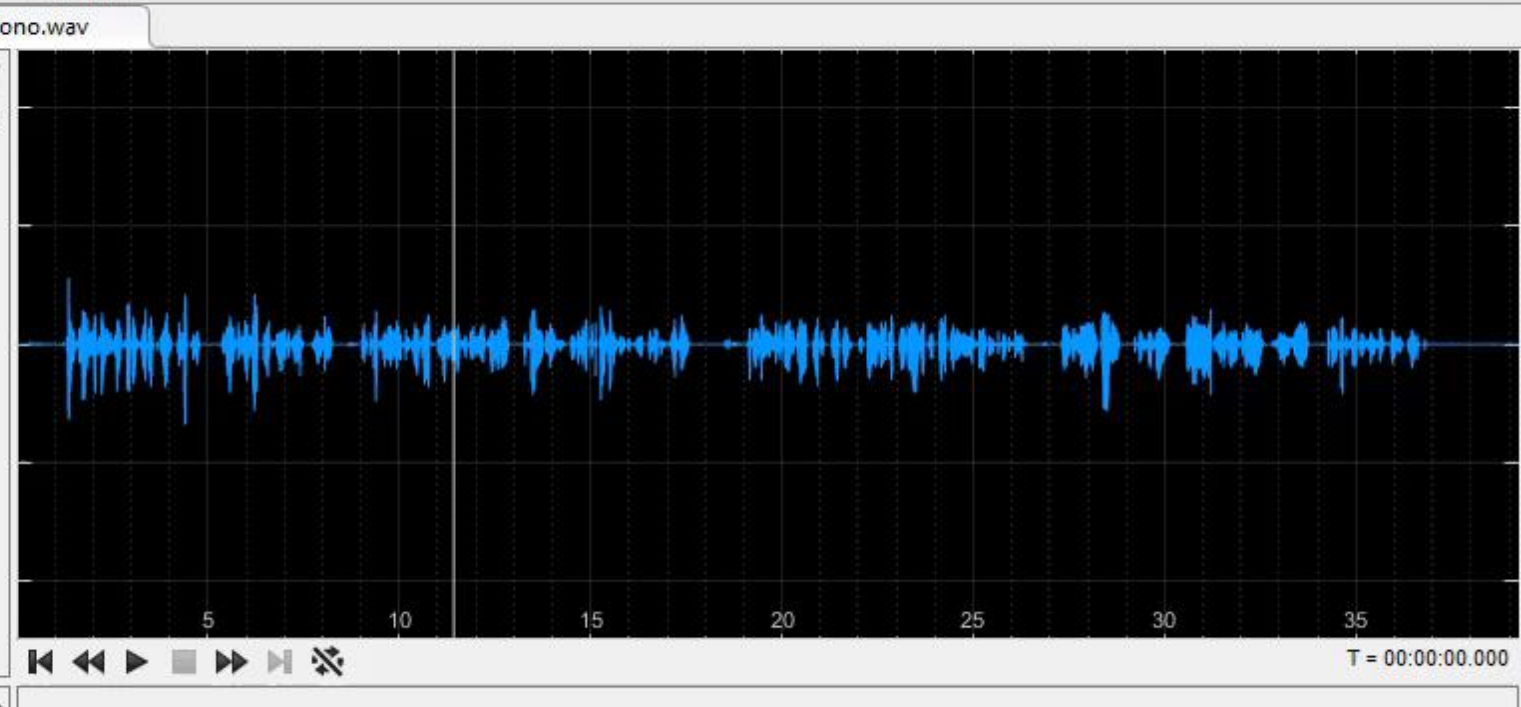
KeywordSpeech-16-16-mono-34secs.flac

ExplainingDetectionRequirements-16-mono.wav

ExplainingDetectionRequirements-16-mono.wav

File Labels + -

To label an audio file, you must first import or add a file label definition.



▼ Audio File Info

ExplainingDetectionRequirements-16-

Channels: 1

Sample Rate: 16000 Hz

Duration: 39.260 s

Compression: Uncompressed

Bit Depth: 16 bits/sample

Location: C:\Docs\Material\Proj

ROI Labels + -

SpeechContent



How to label new non-annotated data?

Use an intelligent system trained to carry out a similar tasks with proven accuracy!

For example:

- Humans
- Pre-trained machine learning models

LABEL | **RECORD** | Cleanup

FILE: Load, Save, Import | DEVICE: Audio Player: Primary Sou..., Settings | VIEW: Default Layout, Legend | AUTOMATION: Speech Detector, Speech to Text | EXPORT: Export

Data Browser

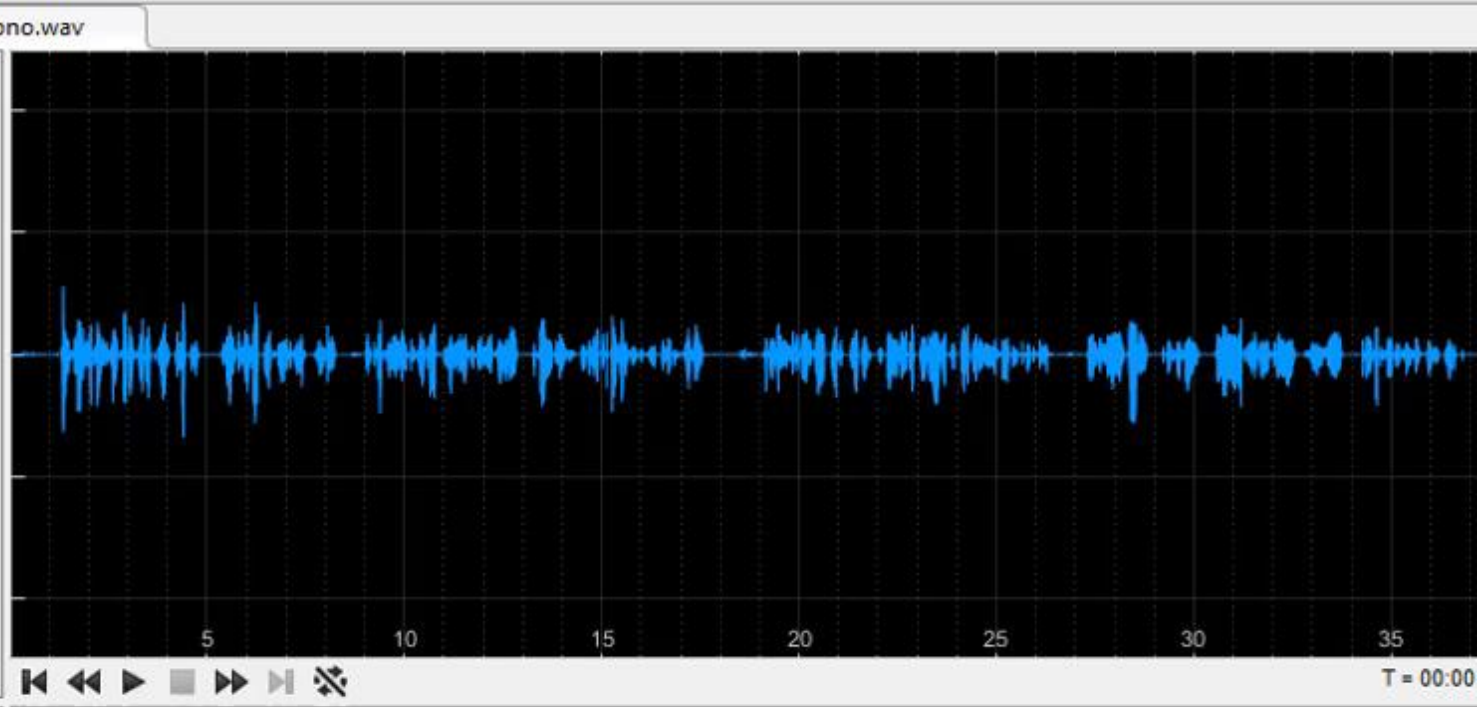
▼ Audio Files

- KeywordSpeech-16-16-mono-34secs.flac
- ExplainingDetectionRequirements-16-mono.wav

ExplainingDetectionRequirements-16-mono.wav

File Labels + -

To label an audio file, you must first import or add a file label definition.



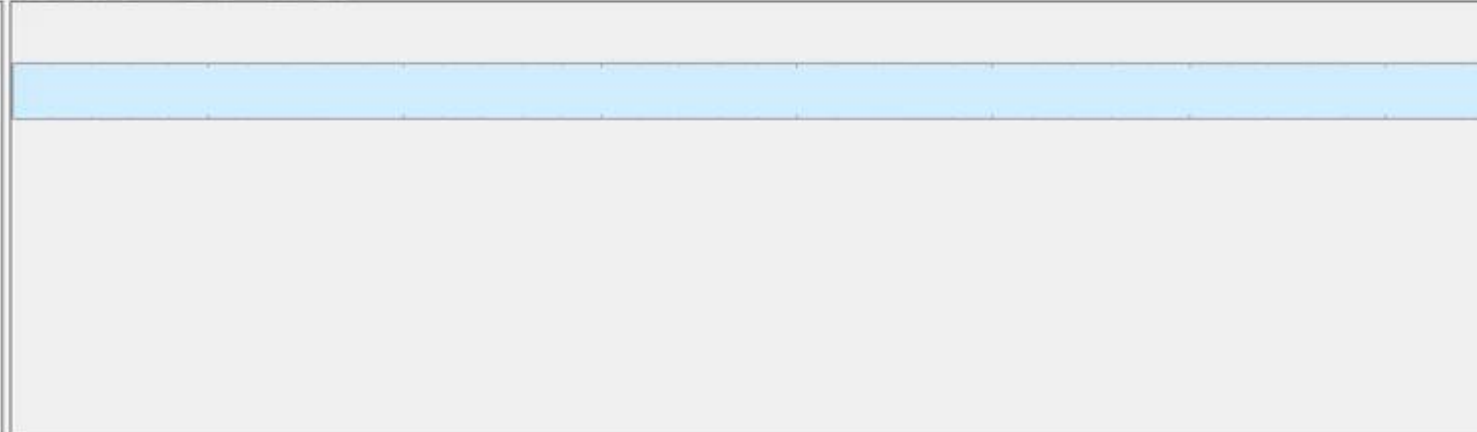
▼ Audio File Info

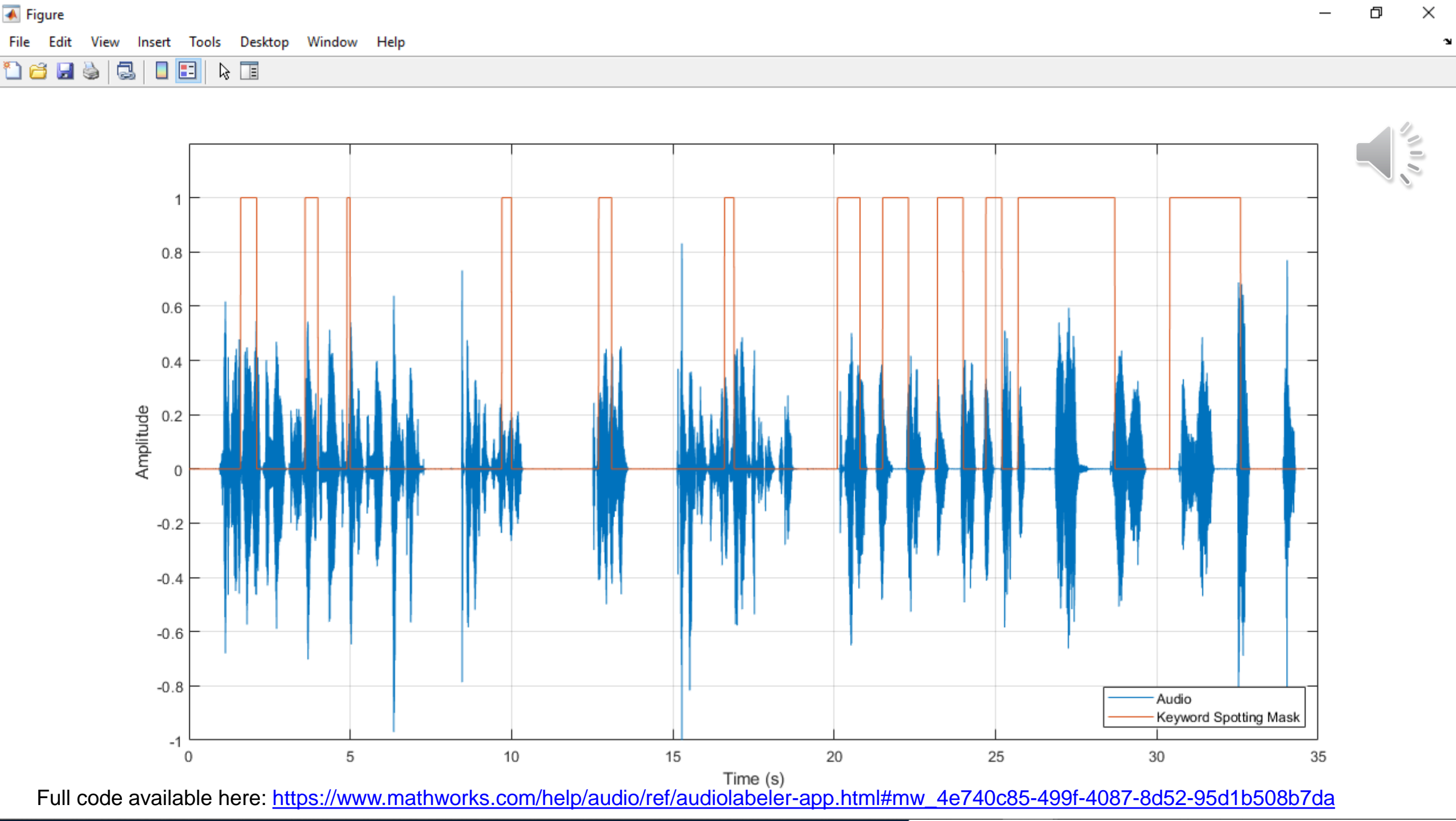
ExplainingDetectionRequirements-16-

Channels: 1
Sample Rate: 16000 Hz
Duration: 39.260 s
Compression: Uncompressed
Bit Depth: 16 bits/sample
Location: C:\Docs\Material\Proj

ROI Labels + -

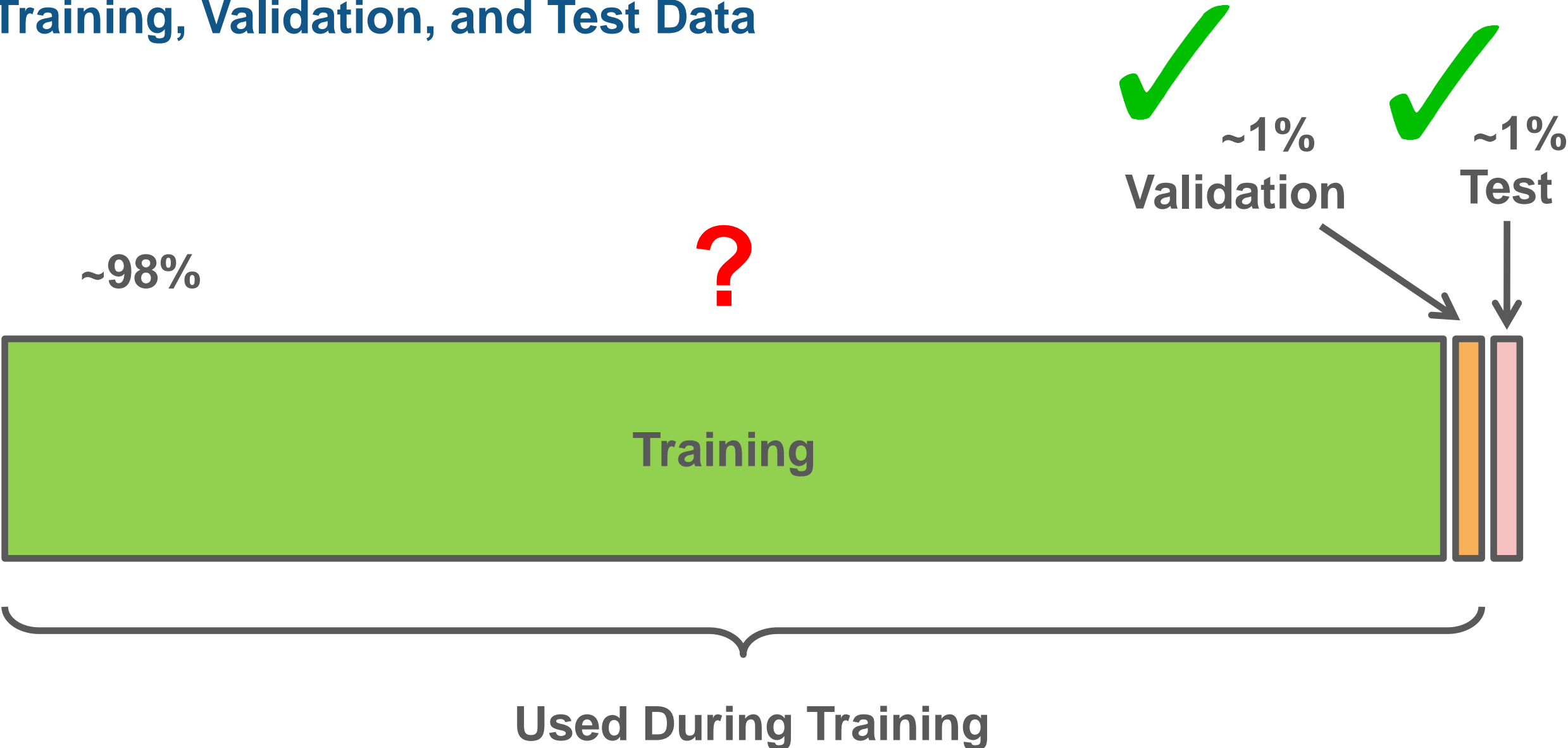
- SpeechContent

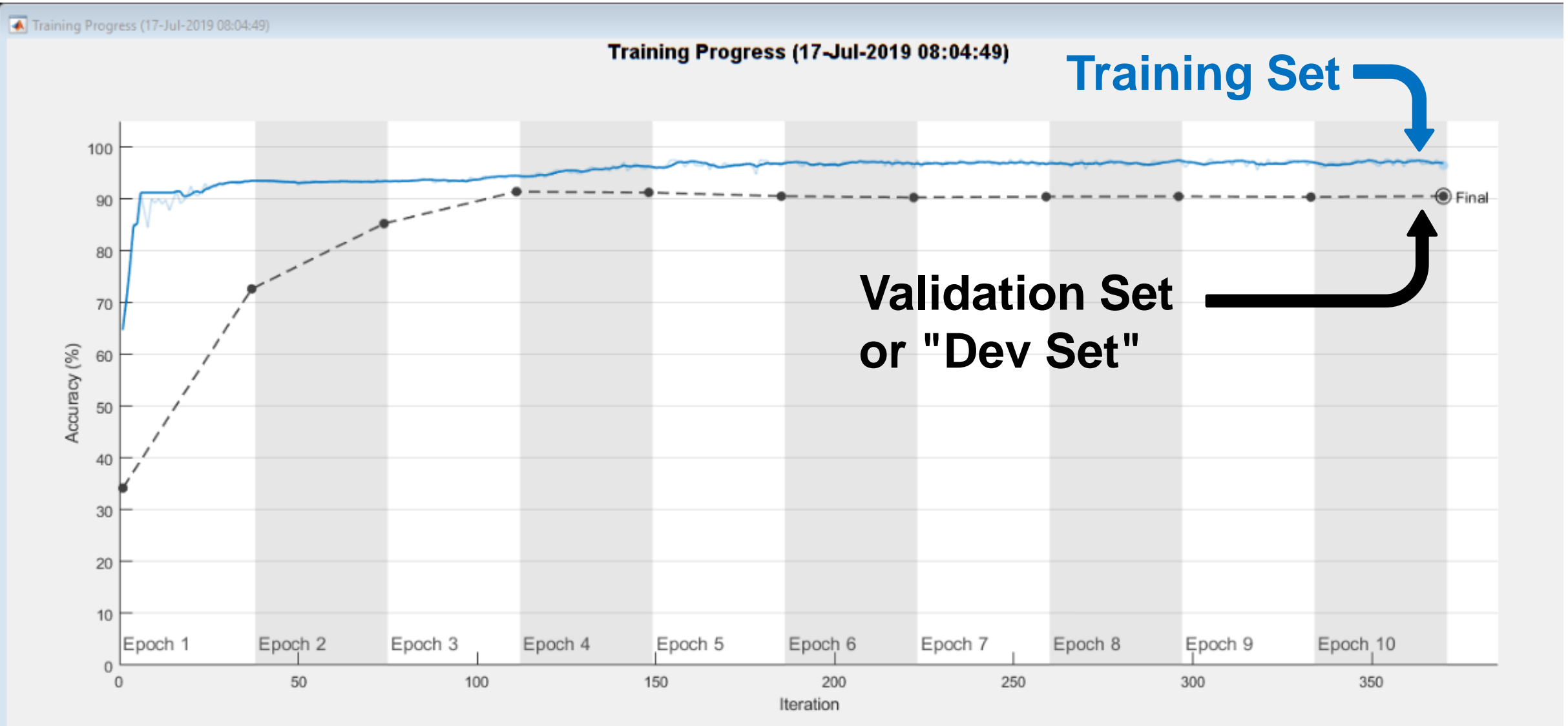




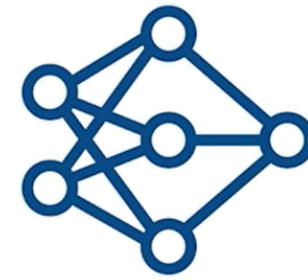
Full code available here: https://www.mathworks.com/help/audio/ref/audiolabeler-app.html#mw_4e740c85-499f-4087-8d52-95d1b508b7da

Training, Validation, and Test Data





Agenda



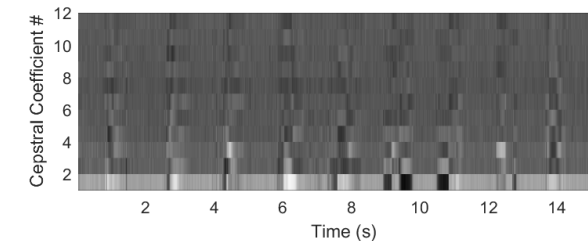
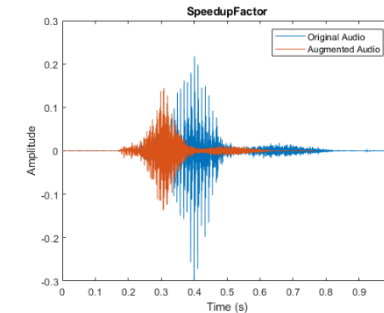
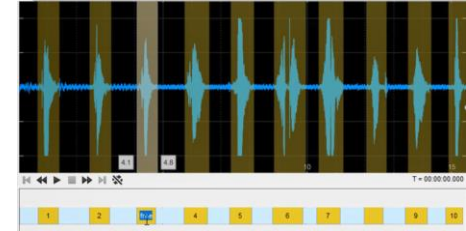
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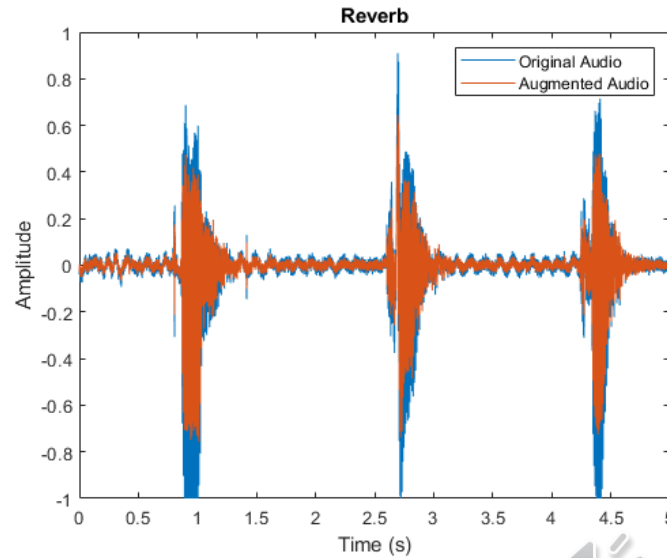


Augmentation – Application-Specific Effects

```
>> auAugm.AugmentationInfo  
ans =  
    struct with fields:
```

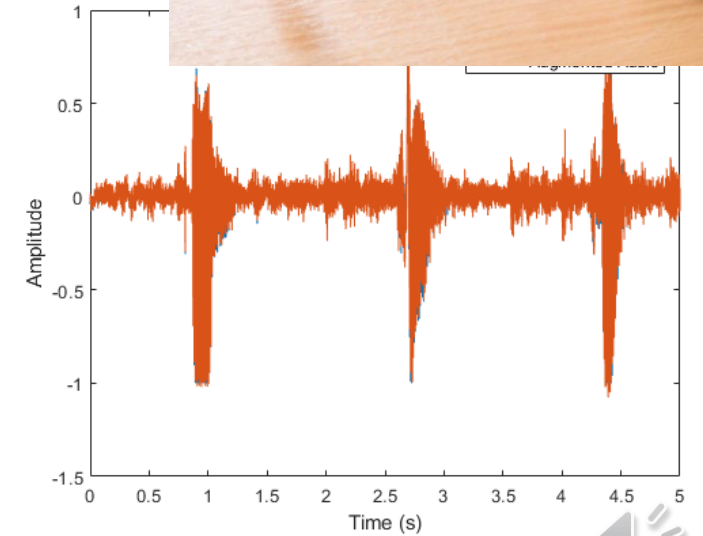


Reverb: 1



Add Kitchen
Reverberation

```
>> a  
ans  
st
```



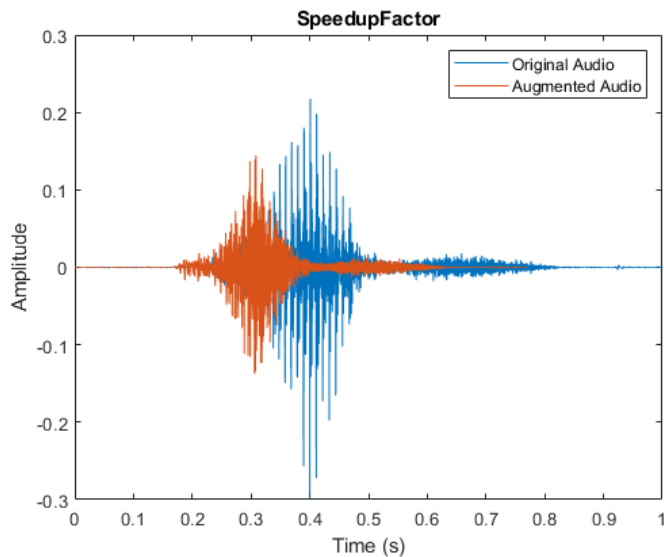
Add Washing
Machine Noise

Augmentation – Common Effective Speech Effects

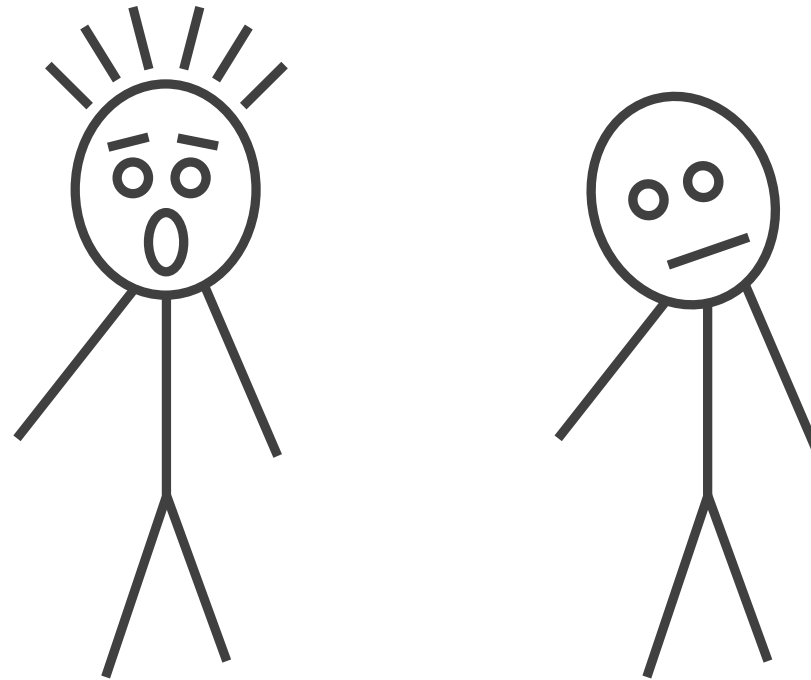


```
>> data.AugmentationInfo(1)
```

```
ans = struct with fields:  
SpeedupFactor: 1.3
```



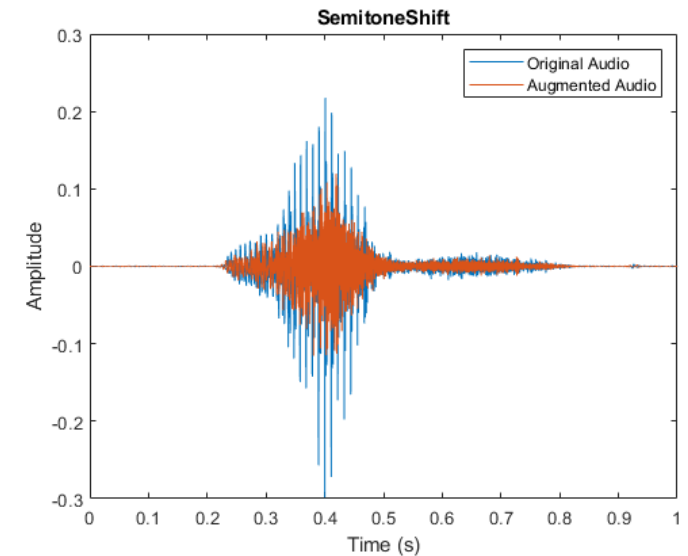
Time
Stretching



Learn more on [audioDataAugmenter](#)

```
>> data.AugmentationInfo(2)
```

```
ans = struct with fields:  
SemitoneShift: -2
```

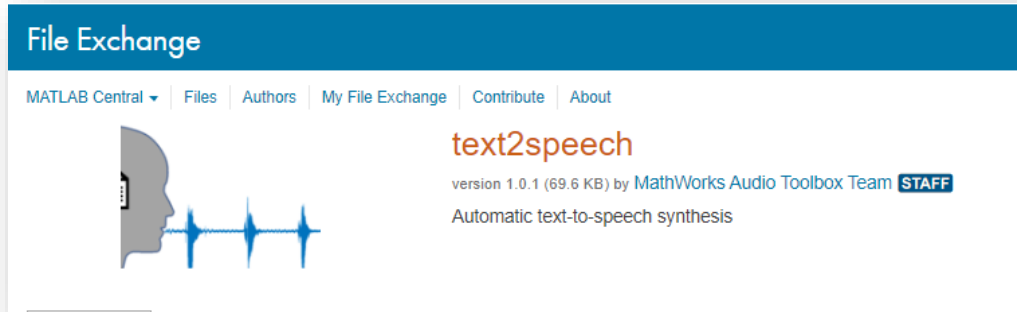


Pitch
Shifting



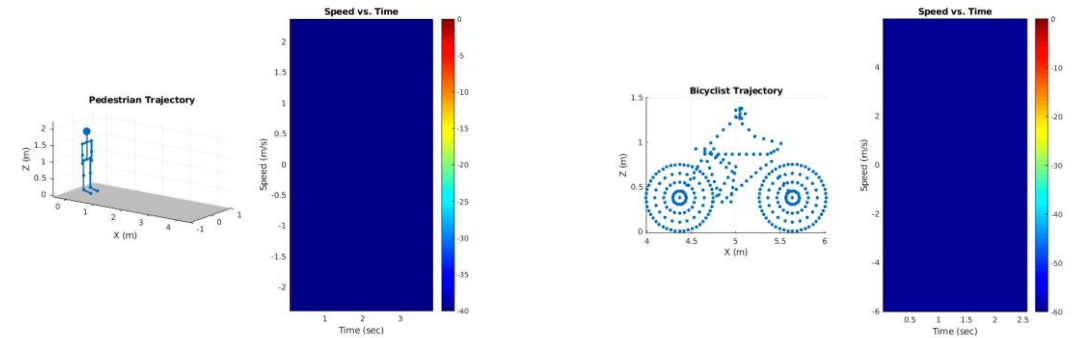
Synthesis – Generative AI models or domain-specific simulations

New text2speech function



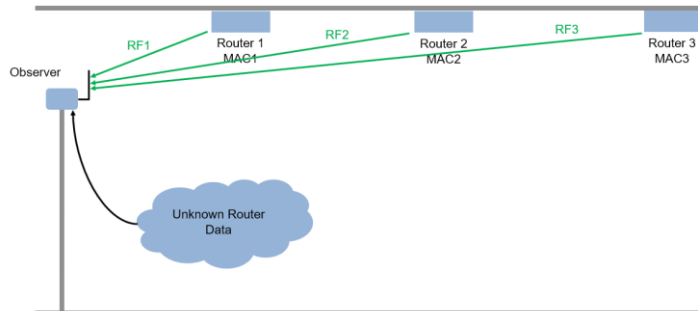
<https://www.mathworks.com/matlabcentral/fileexchange/73326-text2speech>

Pedestrian and Bicyclist (Radar) Classification



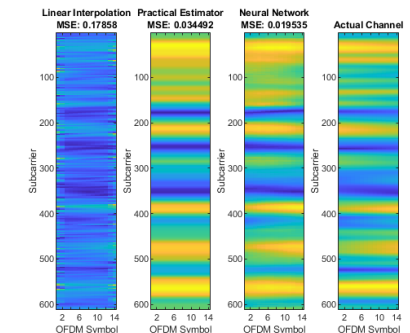
<https://www.mathworks.com/help/phased/examples/pedestrian-and-bicyclist-classification-using-deep-learning.html>

WLAN Router Impersonation Detection



<https://www.mathworks.com/help/comm/examples/design-a-deep-neural-network-with-simulated-data-to-detect-wlan-router-impersonation.html>

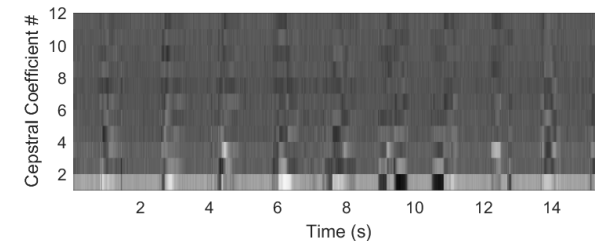
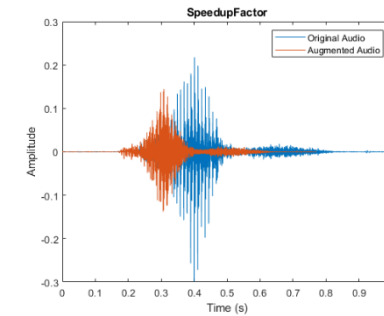
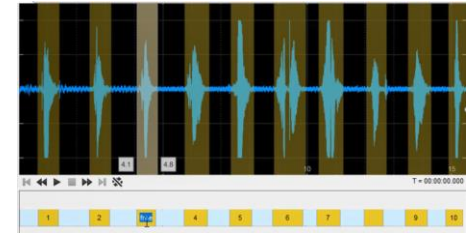
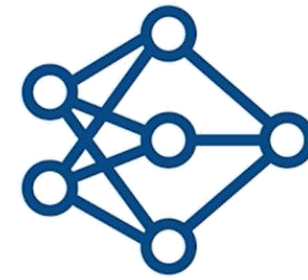
5G Channel Estimation



<https://www.mathworks.com/help/5g/examples/deep-learning-data-synthesis-for-5g-channel-estimation.html>

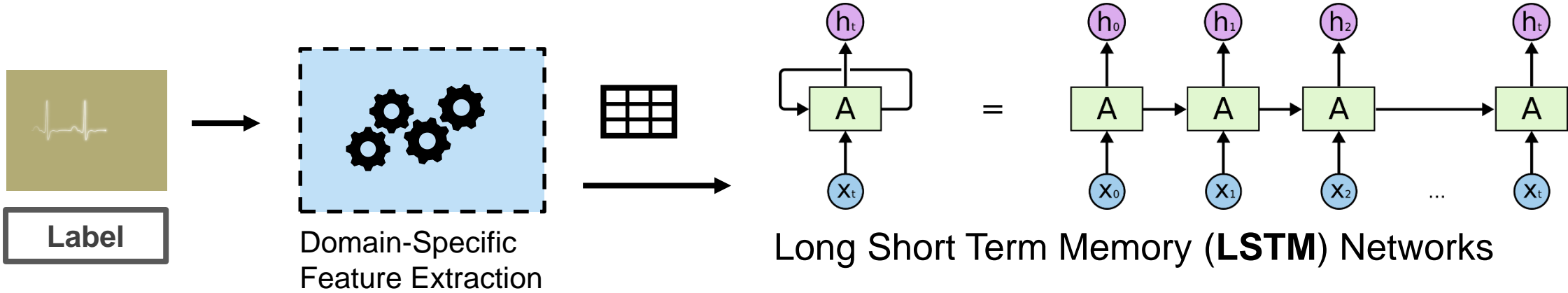
Agenda

- Basics on training deep neural networks for signals
- Annotating data to train networks for practical applications
- Generating new data – synthesis and augmentation
- Creating inputs for deep networks
- From system models to real-time prototypes



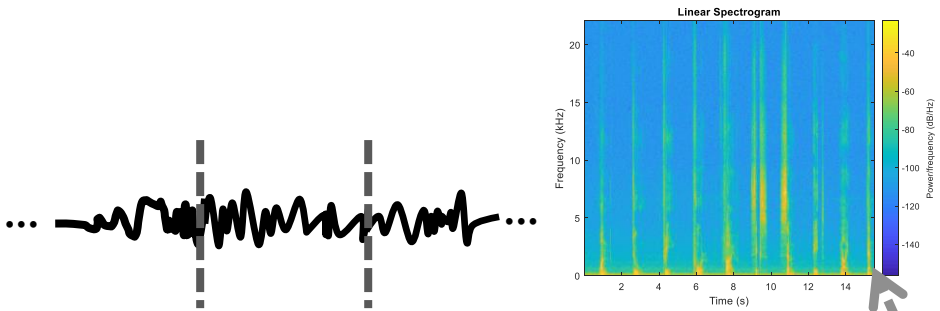
Training deep networks with time-domain signals most often requires extracting features

Deep learning \neq End-to-end learning



Different applications require different feature extraction techniques

spectrogram, stft



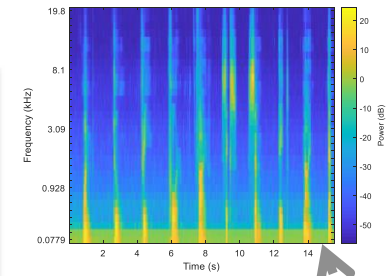
Speech Signal



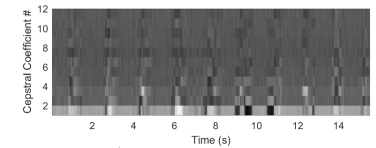
Mel Filterbank



melSpectrogram



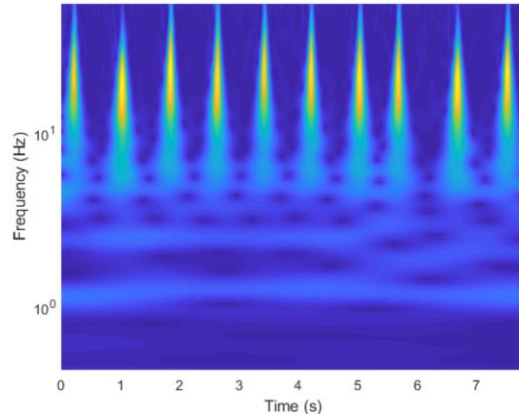
mfcc



Many other time-frequency transforms and signal features

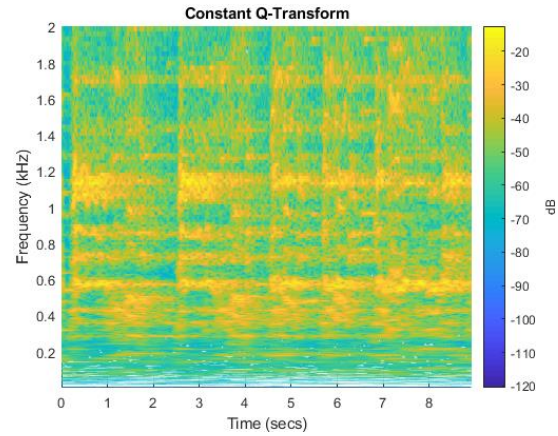
cwt

(Continuous wavelet transform)



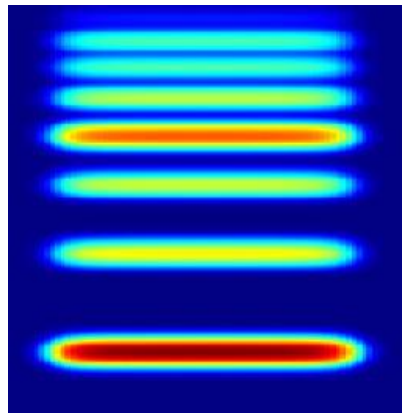
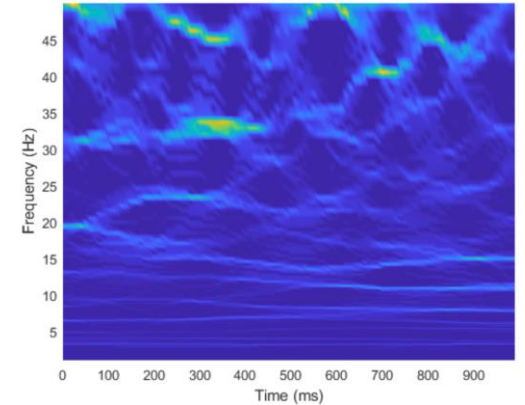
cqt

(Constant Q transform)

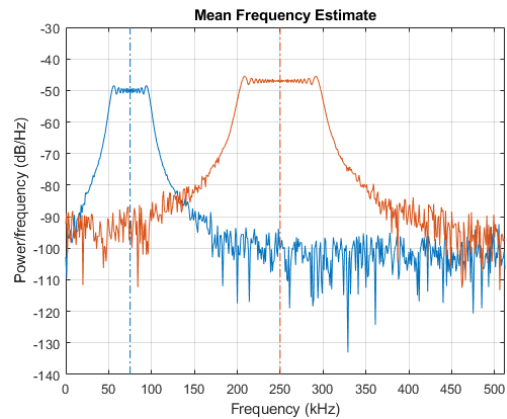


wsstridge

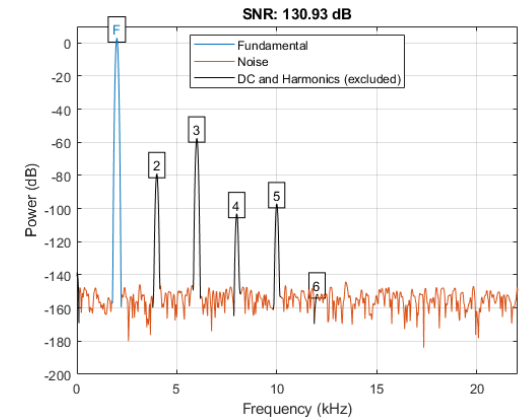
(Synchrosqueezing)



waveletScattering

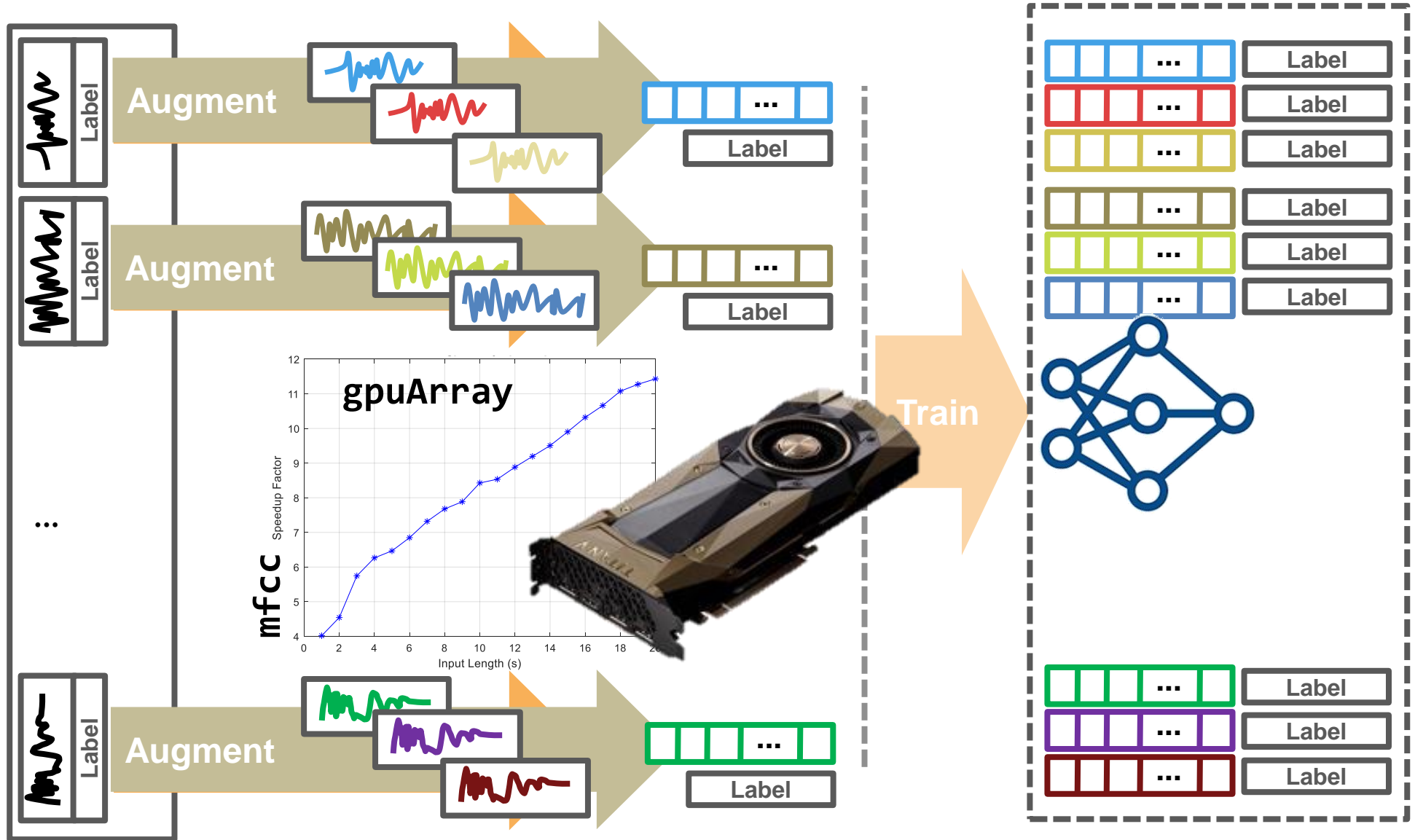


(Spectral statistics)

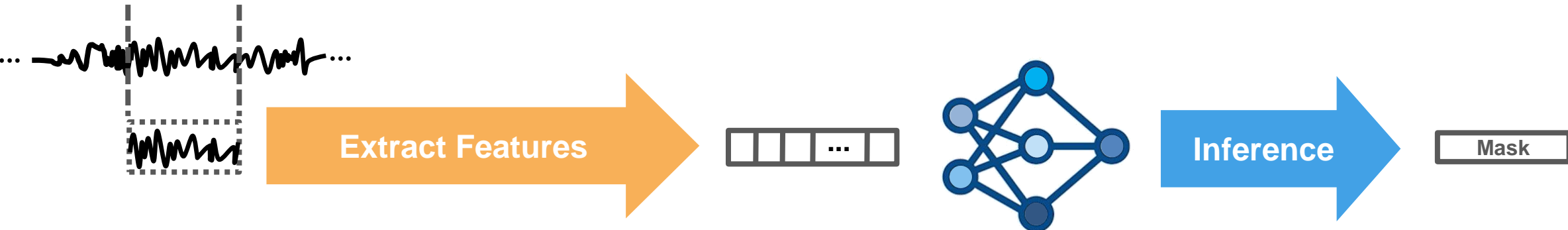


(Harmonic analysis)

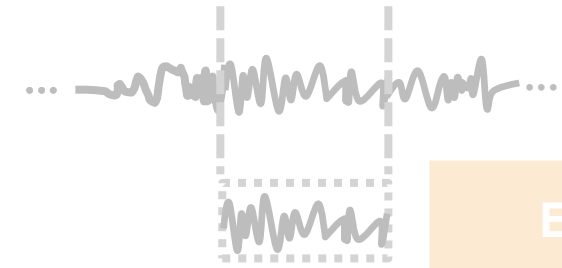
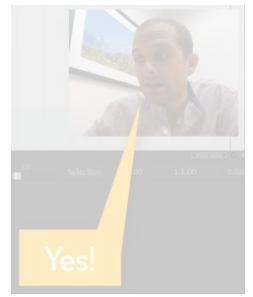
Providing Input Data for Network Training



Using Network for Prediction (aka Inference)



Using Network for Prediction (aka Inference)

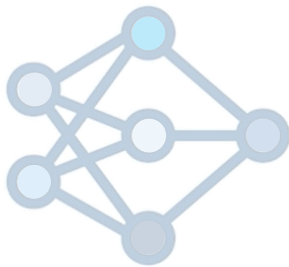


Extract Features

[...]

```
% Extract MFCC from whole analysis buffer  
[coeffs,delta,deltaDelta] = mfcc(buf,SampleRate,...  
    'WindowLength',winLength,...  
    'OverlapLength',ovlpLength);
```

```
% Concatenate and normalize features  
featureMatrix = [coeffs,delta,deltaDelta];  
featureMatrix = (featureMatrix - M)./S;
```



Inference

```
% Detect keyword with LSTM network (Mask around speech keyword)  
featMask = classify(net,featureMatrix.);
```

```
% Debounce and re-align detections in time domain  
[timeMask, chimePosition] = debounceAnalyzeDetectionMask(featMask);
```

```
% Generate chimes for detection events  
chime = generateChimeAtSample(chimePosition,...
```

[...]

Mask

Trigger

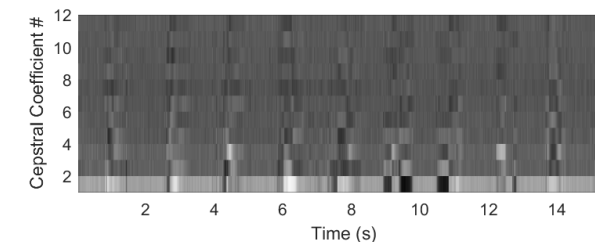
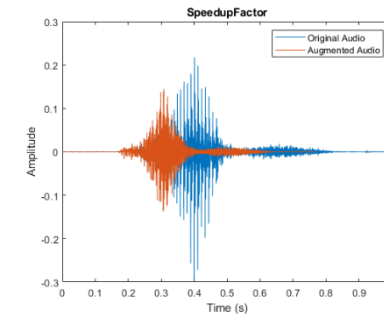
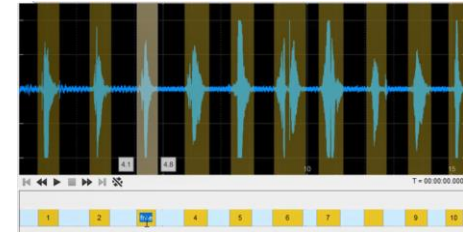
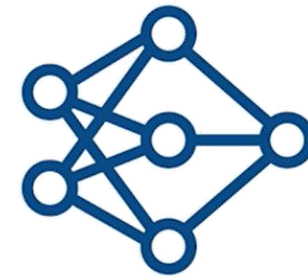


Agenda

- Basics on training deep neural networks for signals
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- Generating new data – synthesis and augmentation

▪ Creating inputs for deep networks

▪ From system models to real-time prototypes



CREATE AND ACCESS DATASETS

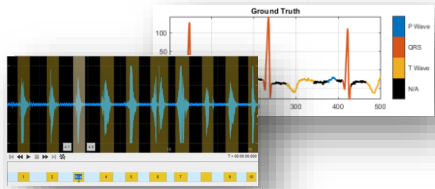
Data sources



Simulation and augmentation

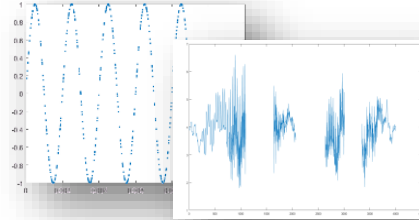


Data Labeling

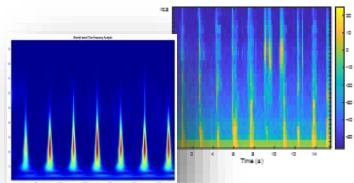


PREPROCESS AND TRANSFORM DATA

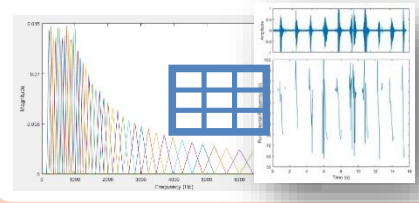
Pre-Processing



Transformation



Feature extraction

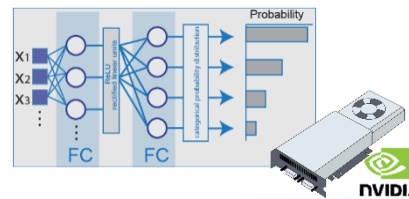


DEVELOP PREDICTIVE MODELS

Import Reference Models/ Design from scratch



Hardware-Accelerated Training

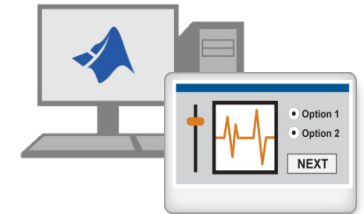


Analyze and tune hyperparameters



ACCELERATE AND DEPLOY

Desktop Apps



Enterprise Scale Systems

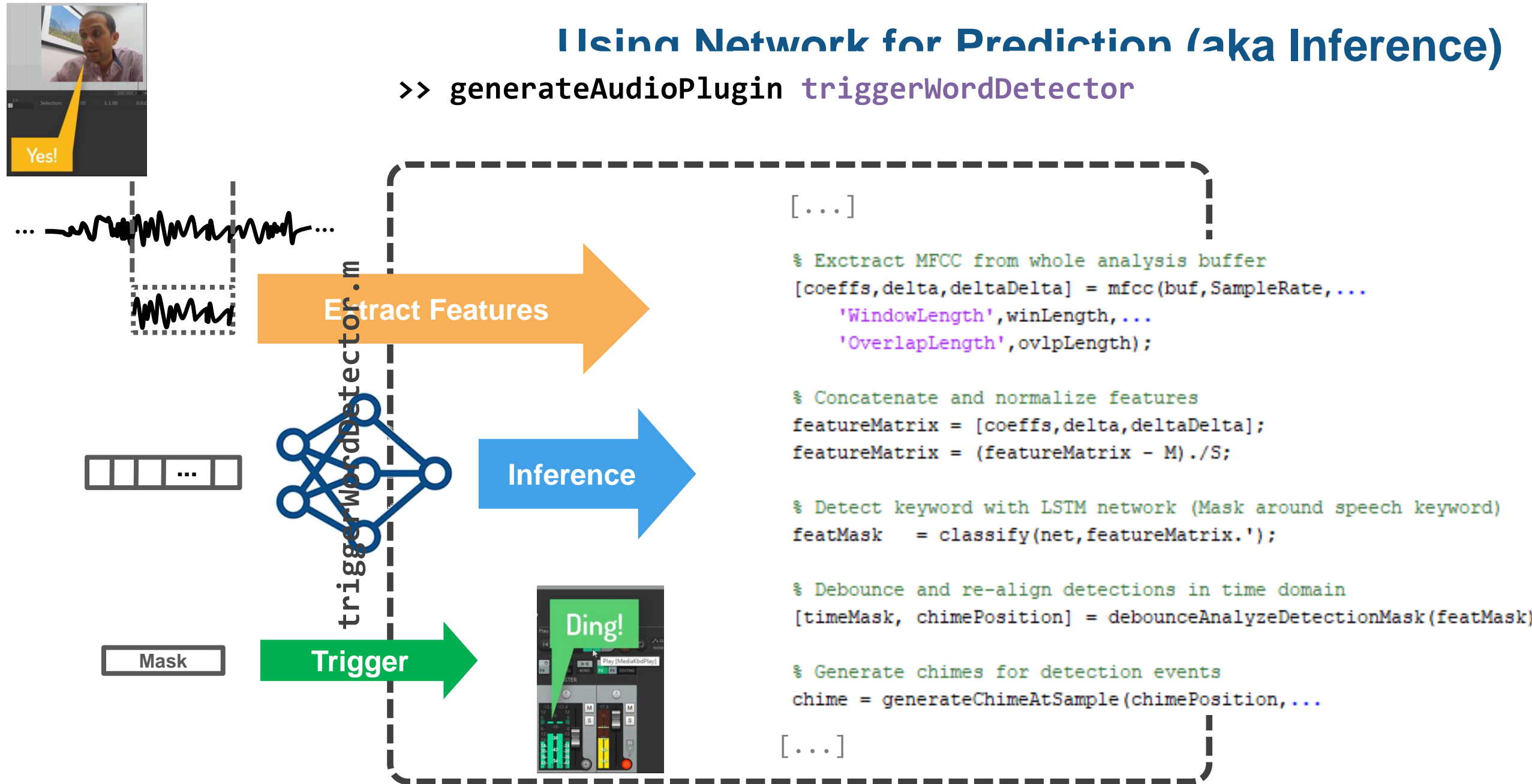
Java
MATLAB
C/C++
Python

Embedded Devices and Hardware



Using Network for Prediction (aka Inference)

>> generateAudioPlugin triggerWordDetector



>> generateAudioPlugin triggerWordDetector

The screenshot shows a code editor window titled "triggerWordDetector_cg - Projucer". The left sidebar displays a file explorer with the following structure:

- triggerWordDetector_cg
- File explorer
- Source
 - triggerWordDetector_cgPluginEditor.h
 - triggerWordDetector_cgPluginEditorResources.h
 - triggerWordDetector_cgPluginProcessor.cpp (highlighted)
 - rt_nonfinite.cpp
 - rtGetNaN.cpp
 - rtGetInf.cpp
 - triggerWordDetector_cg.cpp
 - triggerWordDetector_cg_emxutil.cpp
 - rt_nonfinite.h
 - rtGetNaN.h
 - rtGetInf.h
 - triggerWordDetector_cg_types.h
 - triggerWordDetector_cg.h
 - triggerWordDetector_cg_emxutil.h
 - rtwtypes.h
 - tmwtypes.h
- Filter...
- Modules
- Exporters

The main editor area shows the code for "triggerWordDetector_cgPluginProcessor.cpp". The code includes headers, defines a listener struct, and implements an audio processor class with parameters for GainDB and ChimeLevel.

```
1 #include "../juceLibraryCode/juceHeader.h"
2
3 #include "triggerWordDetector_cg.h"
4
5 struct onParamChangeListener : AudioProcessorValueTreeState::Listener
6
7     onParamChangeListener triggerWordDetector_cgStackData* sd
8     : SD sd
9
10
11
12
13
14     void parameterChanged(const String& parameterID, float newValue) override
15     {
16         void parameterID;
17         int idx = -1;
18         if (parameterID == "GainDB")
19             idx = 0;
20         else if (parameterID == "ChimeLevel")
21             idx = 1;
22     }
23
24     onParamChangeCImpl SD, idx, static_cast<double> newValue;
25
26
27
28
29
30
31
32
33
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37
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68
69
70
71
```

>> generateAudioPlugin triggerWordDetector

The screenshot shows the REAPER v5.35/x64 interface. The main window displays a video track with a video of a man speaking. A VST plugin window titled "VST: Trigger Word Detector (MathWorks) - Track 1" is open, showing parameters for Gain (9.833 dB) and Chime Level (-9.520 dB). A green speech bubble with the text "Ding!" is positioned over the play button in the transport controls. A yellow speech bubble with the text "Yes!" is positioned over the video track. The interface includes a menu bar, a toolbar, a track list, a video preview window, and a mixer.

TriggerWordTest [modified] - REAPER v5.35/x64 - Registered to MathWorks, Inc. (Commercial license)

File Edit View Insert Item Track Options Actions Help [Toggle Track Record Arming] [16kHz 24bit WAV : 2/2ch 1024spls ~119/499ms DirectSound]

VST: Trigger Word Detector (MathWorks) - Track 1

No preset + Param 2 in 2 out UI

Gain 9.833 dB

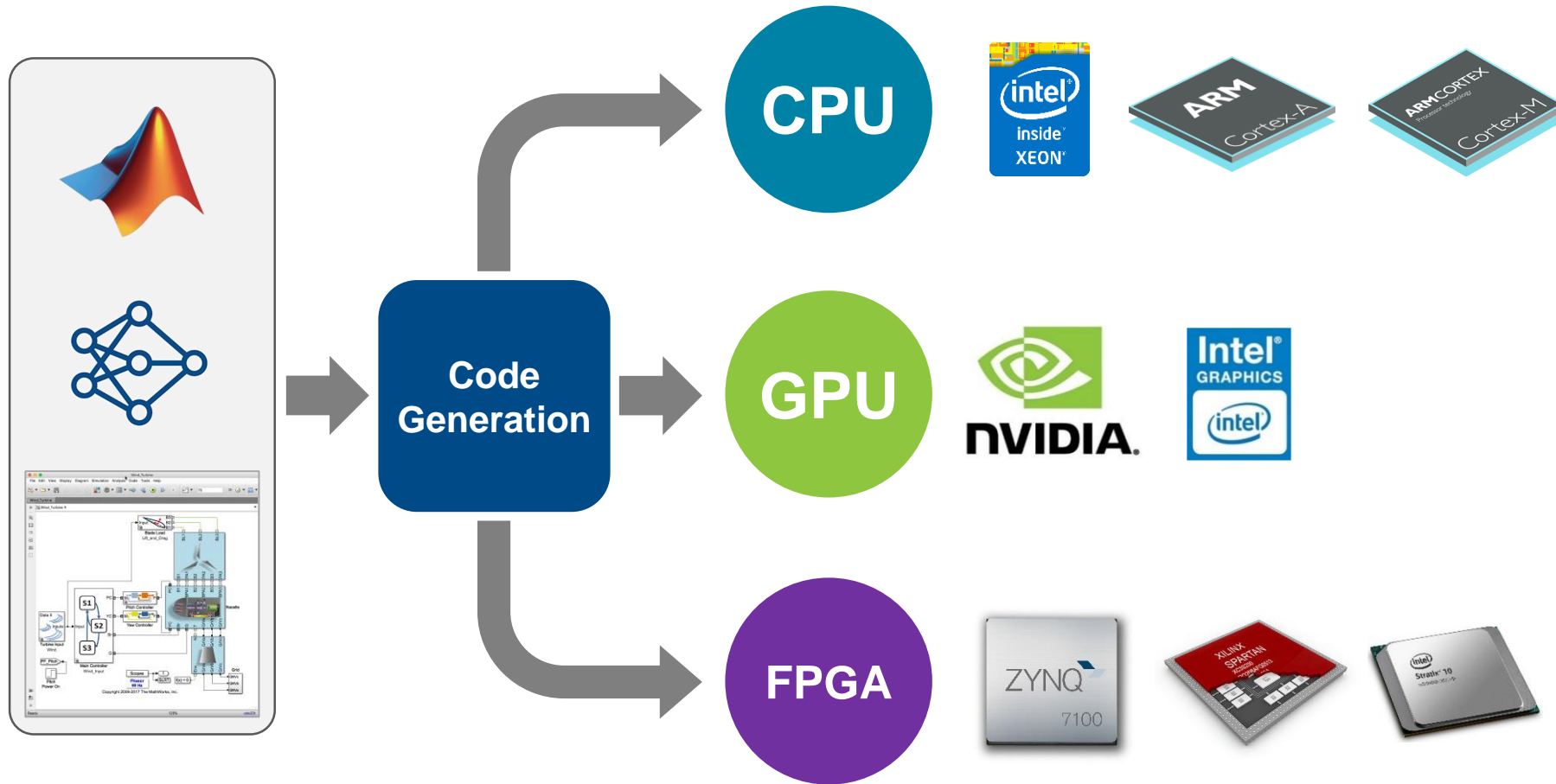
Chime Level -9.520 dB

Ding!

Yes!

Deploy to any processor with best-in-class performance

AI models in MATLAB and Simulink can be deployed on embedded devices, edge devices, enterprise systems, the cloud, or the desktop.



Q: "What do I need to develop such a system?"

A: "A simple and proven deep learning model"

A: "A lot of data, a good dose of signal processing expertise, and the right tools for the specific application in hand"

Deep learning systems can only be as good as the data used to train them

mathworks.com

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