

# MATLAB EXPO 2018

## Riutilizzo e prototipazione di codice

Design of Voice Interfaces for IoT Devices

Francesca Perino



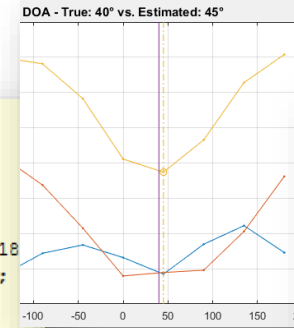
- Innovate

```
% Direction to estimate
incidentAngle = [40;0];

% Theta increments
thetaStep = 45;
steeringThetas = -180:thetaStep:180;
steeringAngles = [steeringThetas;

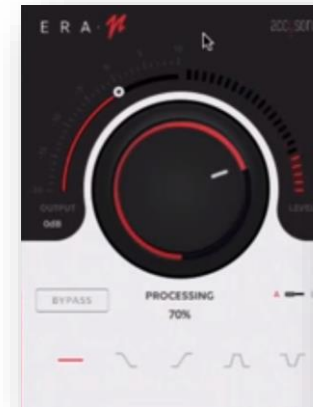
frameSize = 512;

s = beamformArraysSpecificationsDOA(
```

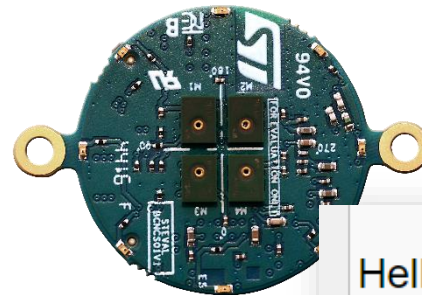


Dataset Recorder	
Rec	Play
Recording Subfolder	H1 Gab BlueCoin Close
SpeechContent	the birch canoe slid on the
SpeakerName	Gabriele Bunkheila
SpeakerGender	Male
SpeakerAge	41

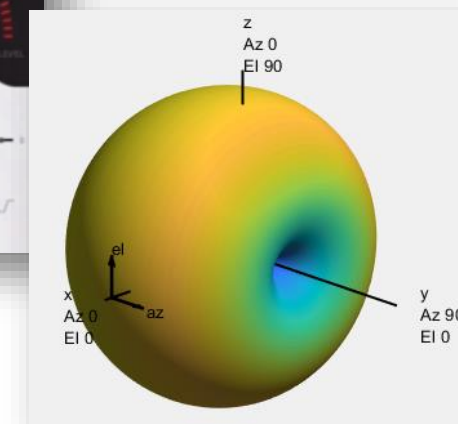
- Reuse



- Prototype



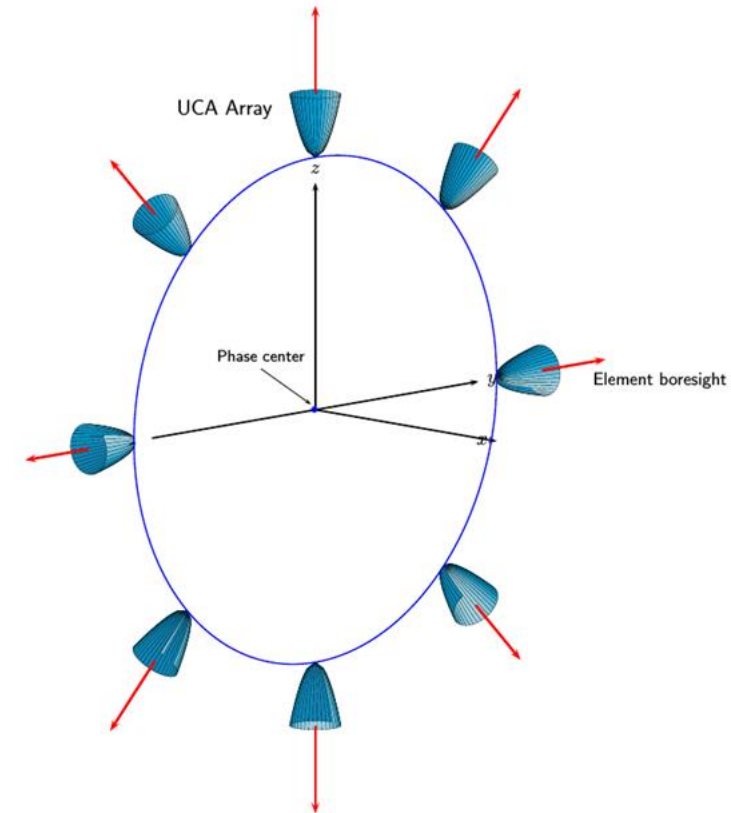
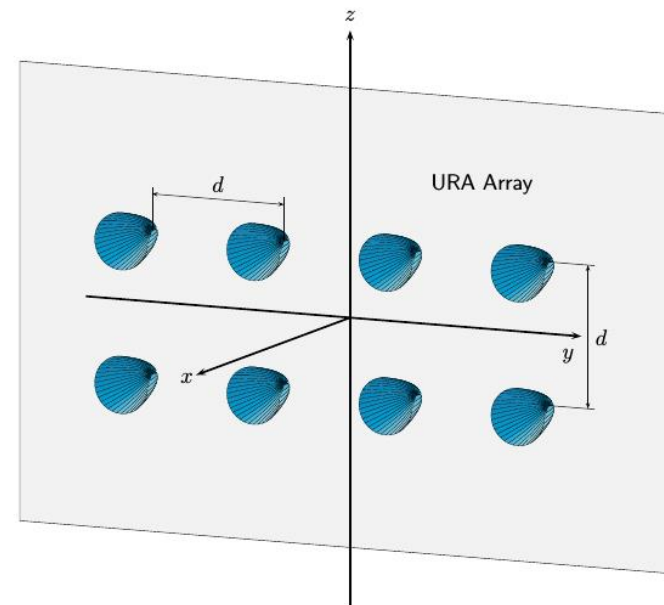
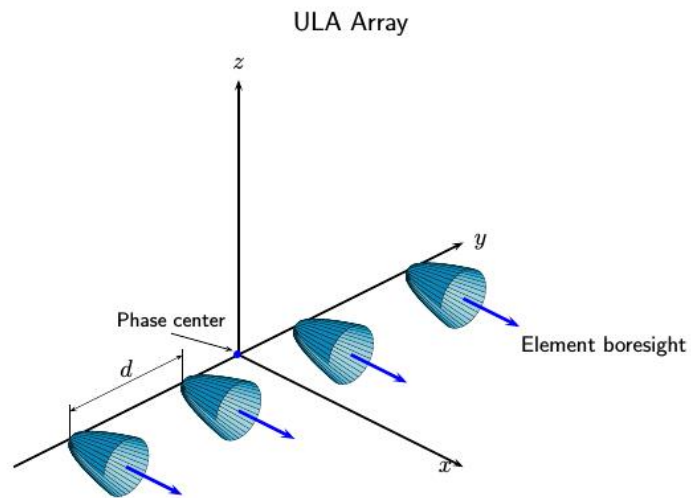
Hello  
Anyone out there



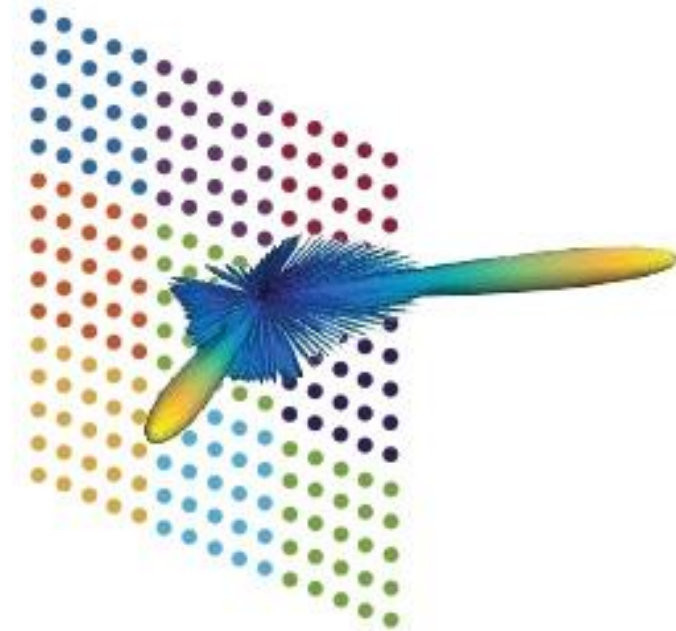
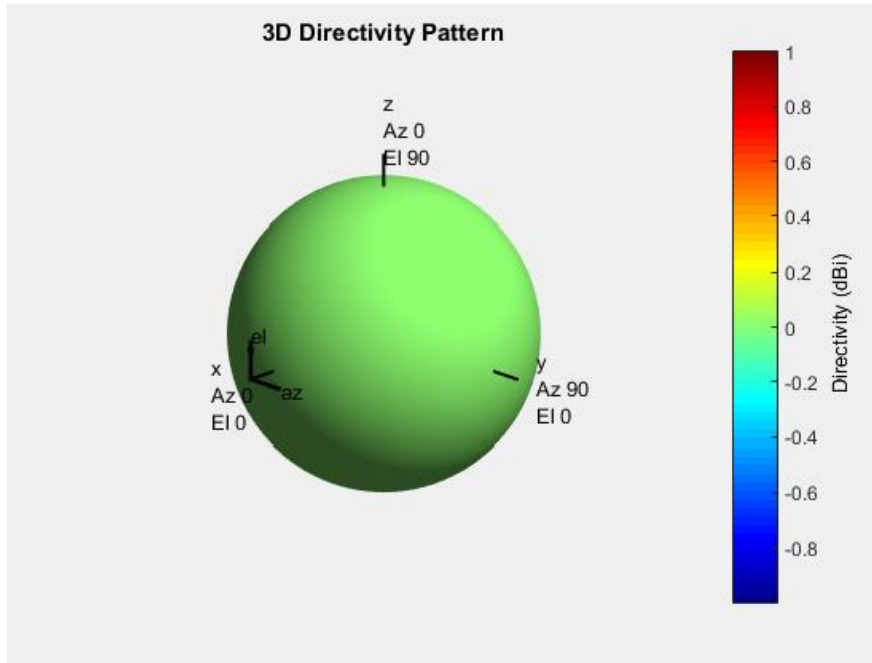
# What Device Is This?



# What Are Microphone Arrays?

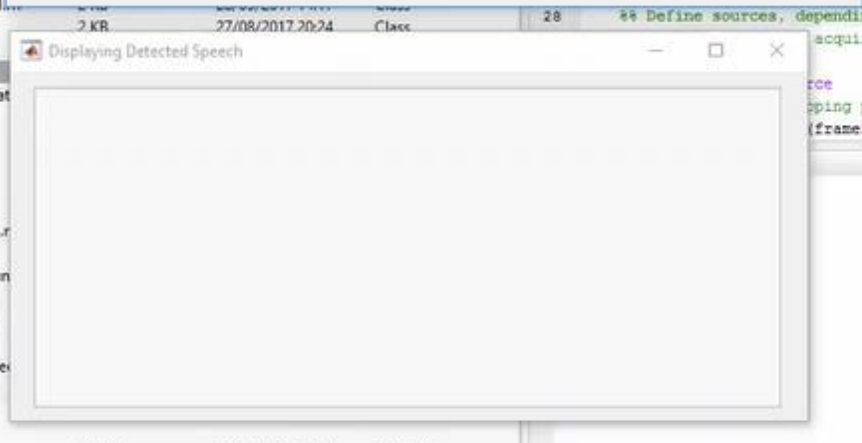
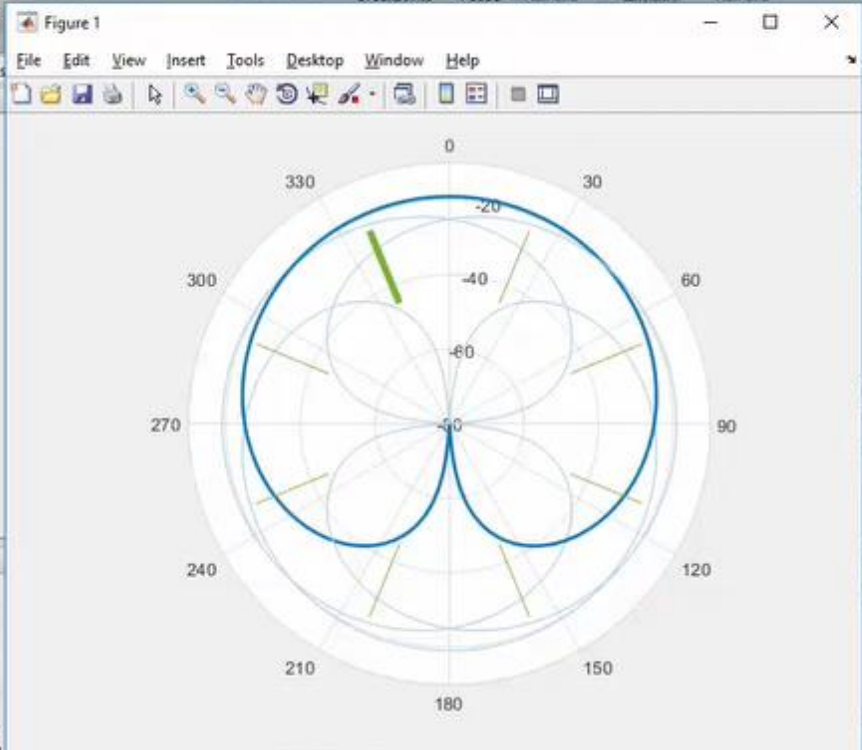


# Why Microphone Arrays?









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Run and Advance

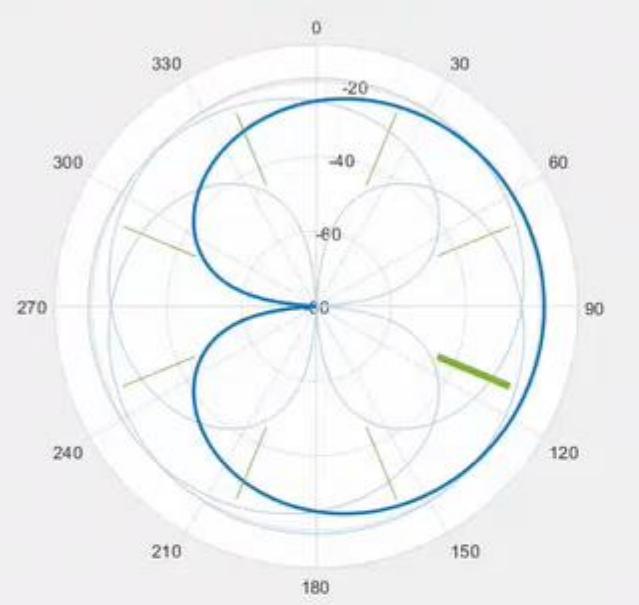
Run and

Figure 1

File Edit View Insert Tools Desktop Window Help

Workspace

Name Value



Current Folder

Application extension

ERA-N.dll

ERA-R.dll

Class

arrayViewer.m

audioArrayParameters.m

displayStreamingTextInUI.m

ScrollingStringList.m

Function

adjustPositions.m

beamformArraysSpecificat

bufferAndSearch.m

bwlabels1.m

bwregionspans.m

hoppingSource.m

levelChannels.m

selectArrayConfiguration.r

tuneBypassDSP.m

tuneNoiseRemovalAmon

Script

PostToSpeechDisplay.m

PrototypeSystem.m

TestSegmentationForSpee

MAT-file

MicGainRatios.mat

WAV File

onetwothreetest.wav

2 KB

27/08/2017 20:24

C:\class

28

% Define sources, depending

acquisit

nce

pping par

(frameSiz

Displaying Detected Speech

good afternoon everybody

Time Scope

File Tools View Playback Help

Amplitude

0.5

0.4

0.3

0.2

0.1

0


-0.1

-0.2

-0.3

-0.4

-0.5



Time (secs)

0

1

2

3

4

5

6

7

8

9

10

Processing

T=0

UK EXPO\Talk PPT\Final V

transforming through inv

blem parameters

fications(fs);

rays);

ech recognition

l\Projects\2017-xx -

ffer = 2;

ech

Buf

01

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08

09

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100

adjustPositions.m (Function)

Busy

script

Ln 4 Col 15



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Go To

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fx

Breakpoints

Pause

Run and Advance

Run and

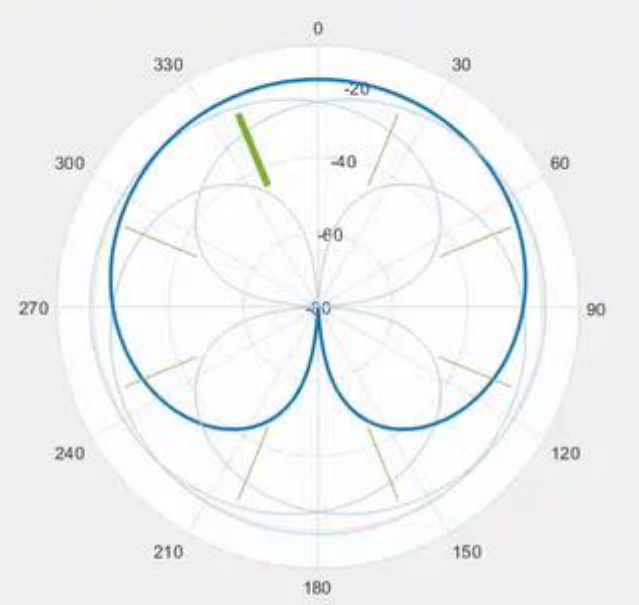
Figure 1

File Edit View Insert Tools Desktop Window Help

FILE

Workspace

Name Value



Current Folder

Application extension

Class

Function

Script

MAT-file

WAV File

adjustPositions.m (Function)

Busy

UK EXPO\Talk PPT\Final V

transforming through inv

blem parameters

fications(fs);

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ech recognition

l\Projects\2017-xx -

ffer = 2;

rch

Buf

tre

Time Scope

File Tools View Playback Help



Processing

T=2

Displaying Detected Speech

good afternoon everybody  
do you speak Matlab

2 KR

27/08/2017 20:24

Class

28

% Define sources, depending

acquisit

nce

pping par

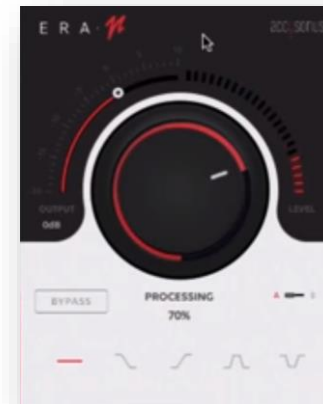
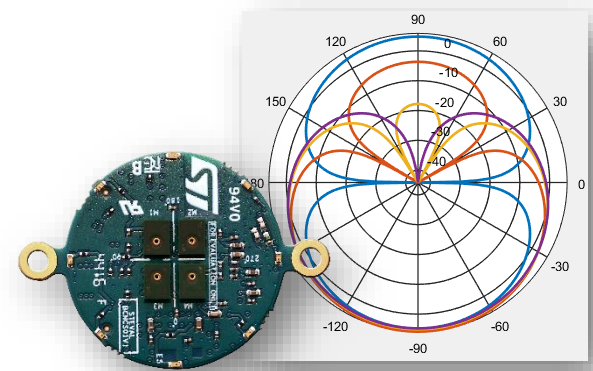
(frameSiz

script

Ln 4 Col 15

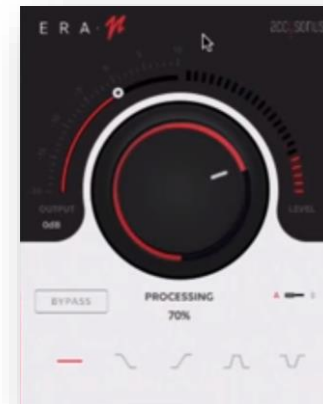
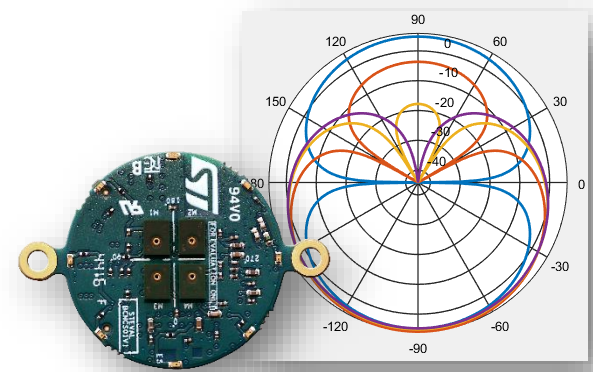
# How Can I...

1. Design a microphone array system?
2. Validate my voice interface can work in real-life scenarios?
3. Understand what else can help me improve my performance?



# How Can I...

1. Design a microphone array system?
2. Validate my voice interface can work in real-life scenarios?
3. Understand what else can help me improve my performance?

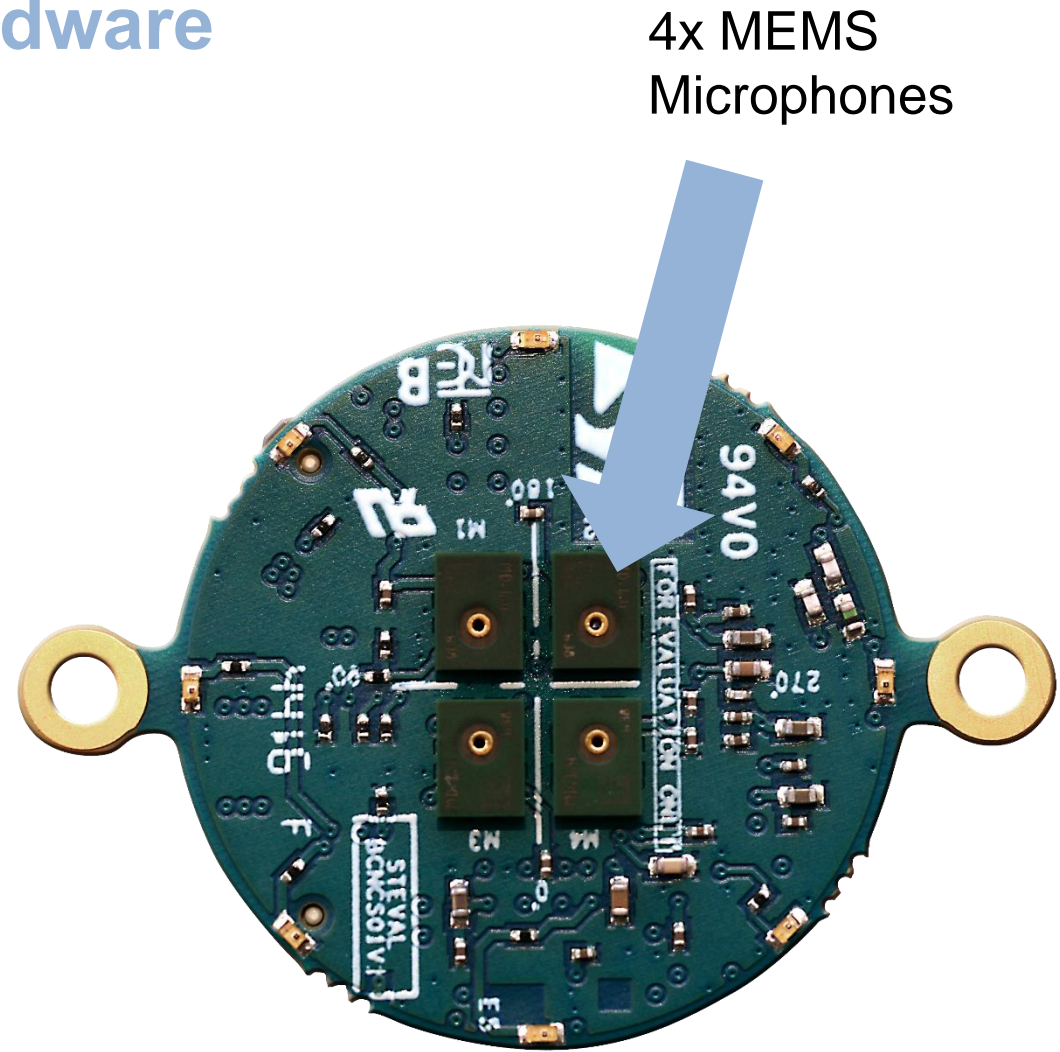
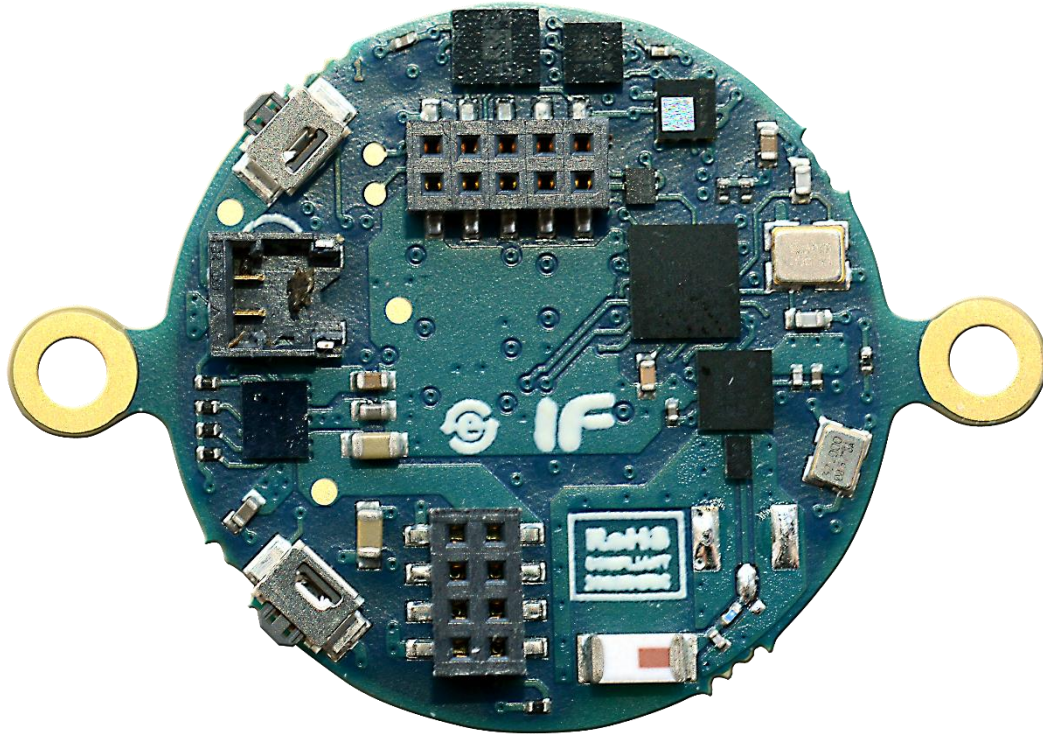


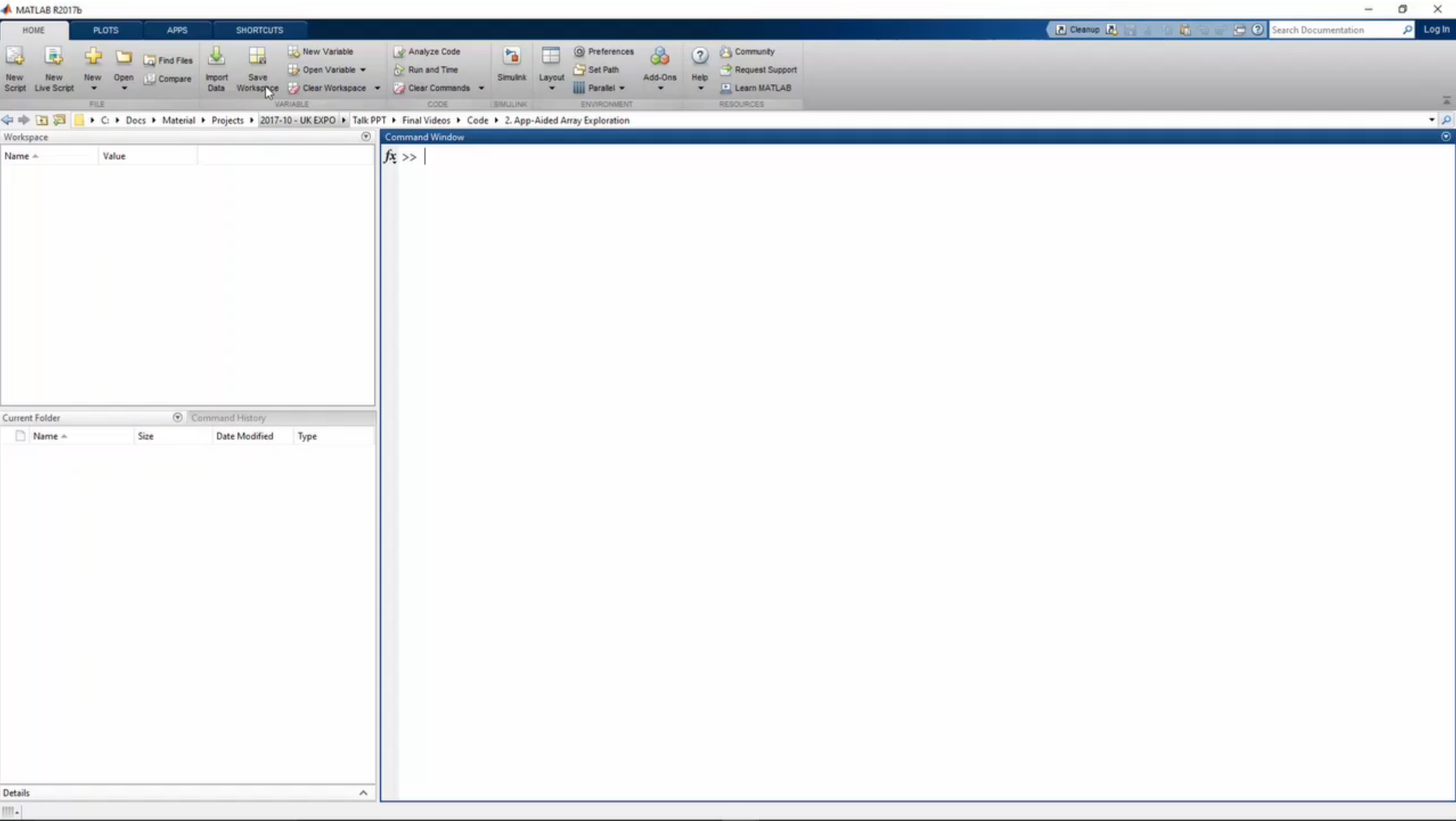


# Design a Microphone Array System

...starting from a given array hardware

- [BlueCoin](#) from ST Microelectronics







### Array Settings

Array Type: Uniform Rectangular

Element Type: Isotropic Antenna

BackBaffled: ☐

Size: [2 2]

Element Spacing: [0.003 0.003]

m

Signal Frequencies: 1000

Hz

Lattice: Rectangular

Array Normal: x

Propagation Speed: 340

m/s

Steering: ☐

Row Taper: None

Column Taper: None

Apply

### Visualization Settings

View: Array Geometry

Show Normals: Array Geometry

Show Index: 2D Array Directivity

3D Array Directivity

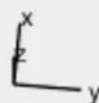
Grating Lobe Diagram

### Array Characteristics

Array Directivity: 0.00 dBi at 0 Az; 0 El

Array Span: x=0 m y=3 mm z=3 mm

Number of Elements: 4





## Array Settings

Array Type: Uniform Linear

Element Type: Isotropic Antenna

BackBaffled: ☐

Number of Elements: 2

Element Spacing: 0.003 m

Array Axis: y

Signal Frequencies: 1000 Hz

Propagation Speed: 340 m/s

Steering: ☐

Taper: None

Apply

## Visualization Settings

View: Array Geometry

Show Normals: ☐Show Index: ☐

## Array Characteristics

Array Directivity: 0.00 dBi at 0 Az; 0 El

Array Span: x=0 m y=3 mm z=0 m

Number of Elements: 2





## ▼ Array Settings

Array Type: Uniform Linear

Element Type: Isotropic Antenna

BackBaffled: ☐

Number of Elements: 2

Element Spacing: 0.003 m

Array Axis: y

Signal Frequencies: 1000 Hz

Propagation Speed: 340 m/s

Steering: ☒

Steering Angles: [90;0] deg

Phase Shift Quantization: 0 bits

Taper: Custom

Custom Taper: [1, -1]

Apply

## ▼ Visualization Settings

View: 2D Array Directivity

Cut Type: Azimuth Cut-Polar

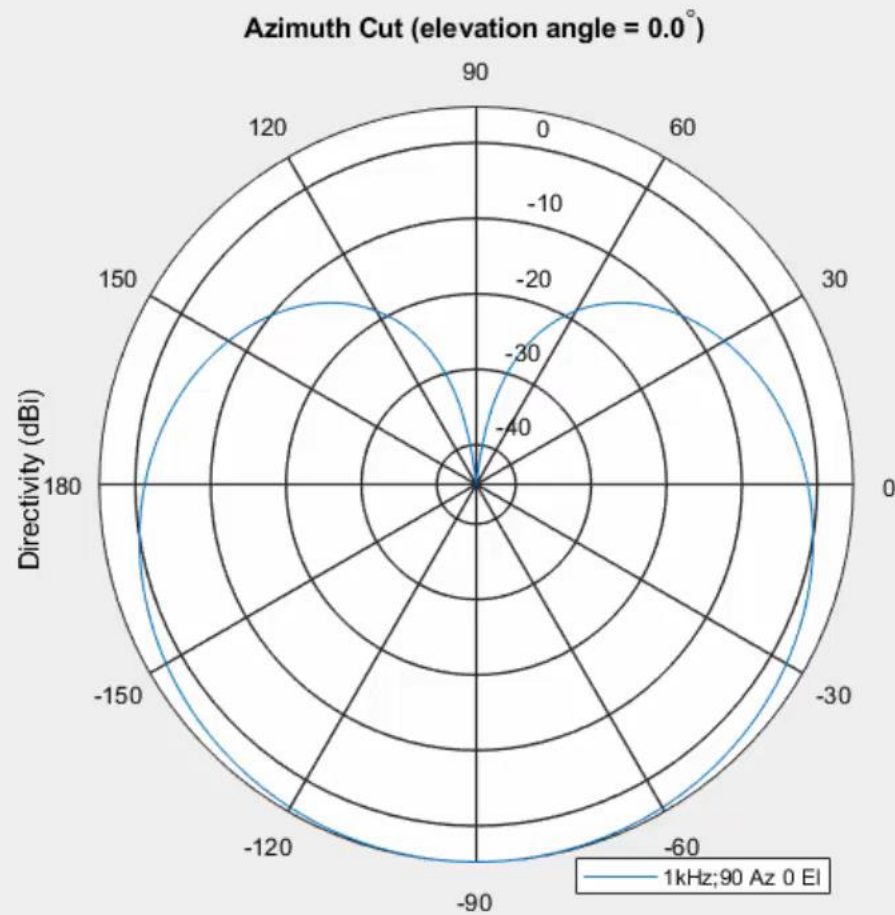
Cut Value: 0 deg

## ▼ Array Characteristics

Array Directivity: -Inf dBi at 90 Az; 0 El

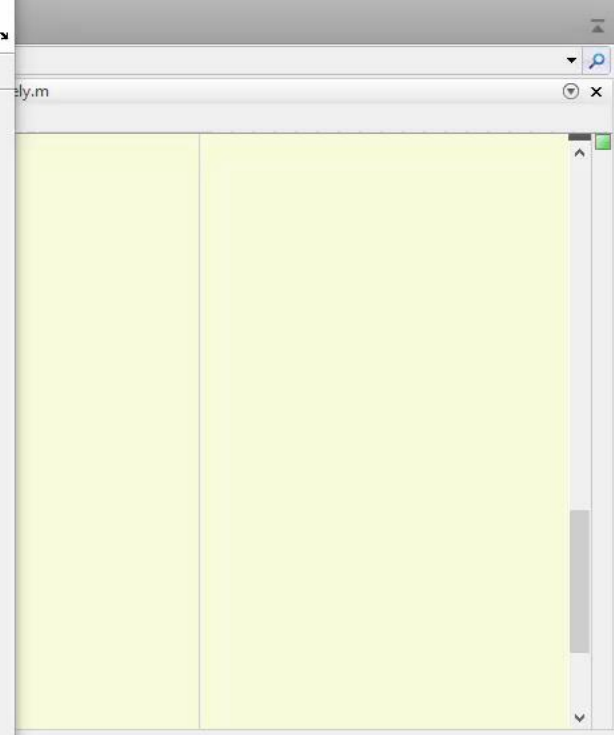
Array Span: x=0 m y=3 mm z=0 m

Number of Elements: 2



Directivity (dBi), Broadside at 0.00 degrees





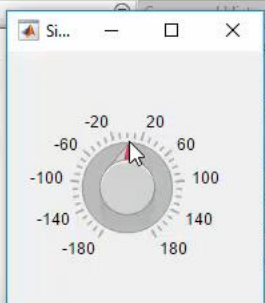
The screenshot shows a MATLAB environment. On the left, a file explorer pane displays the following structure:

- Current Folder**
  - Folder
    - OLD
  - Class
    - arrayPatternViewer.m
    - audioArrayParameters.m
    - pointingArrayViewer.m
    - steeringArrayViewer.m
  - Function
 

File Name	Size	Progress
adjustPositions.m	1 KB	29/
beamformArraysSpecifications.m	5 KB	29/
selectArrayConfiguration.m	1 KB	14/
tuneParameter.m	1 KB	14/
  - Script
 

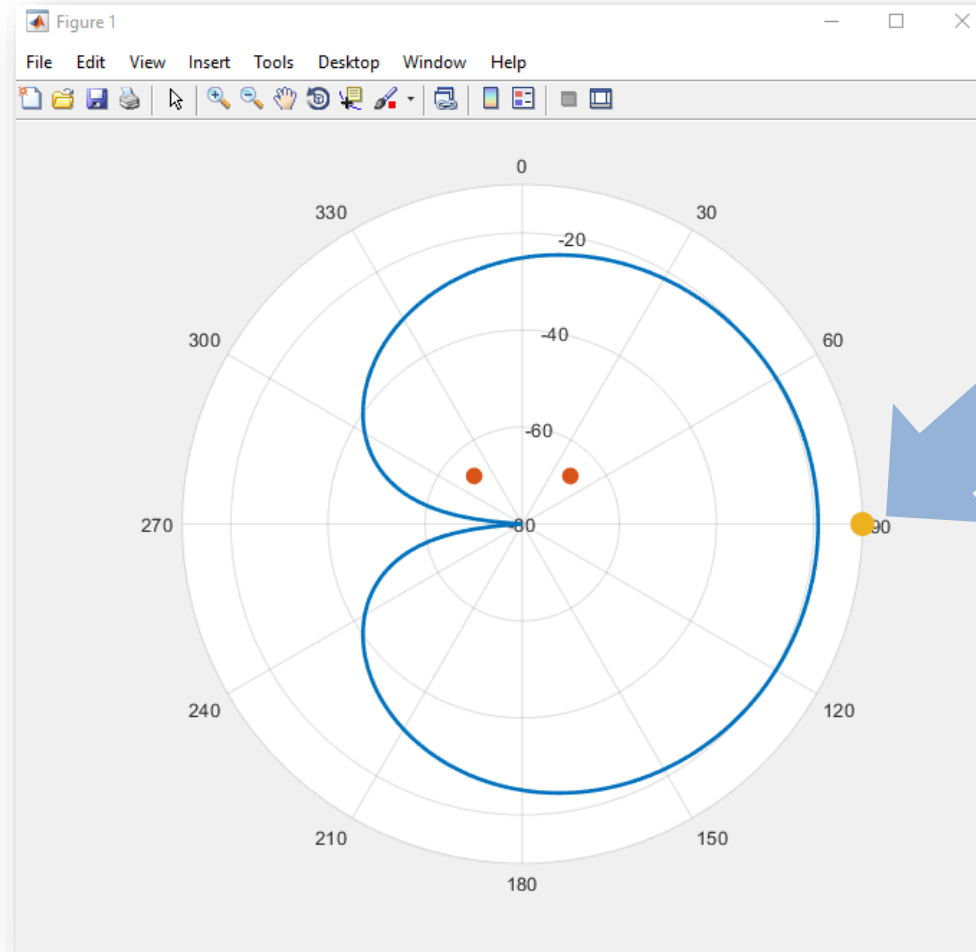
File Name	Size	Progress
SteerInteractively.m	2 KB	29/

On the right, a window titled 'Si...' is open, showing a circular control interface. It features a central knob with a red mouse cursor pointing to it. The knob has numerical markings around its perimeter: -180, -140, -100, -60, -20, 20, 60, 100, 140, and 180. The interface is designed for interactive steering of an array.





# Time Domain Simulation of an Array System



Sound source

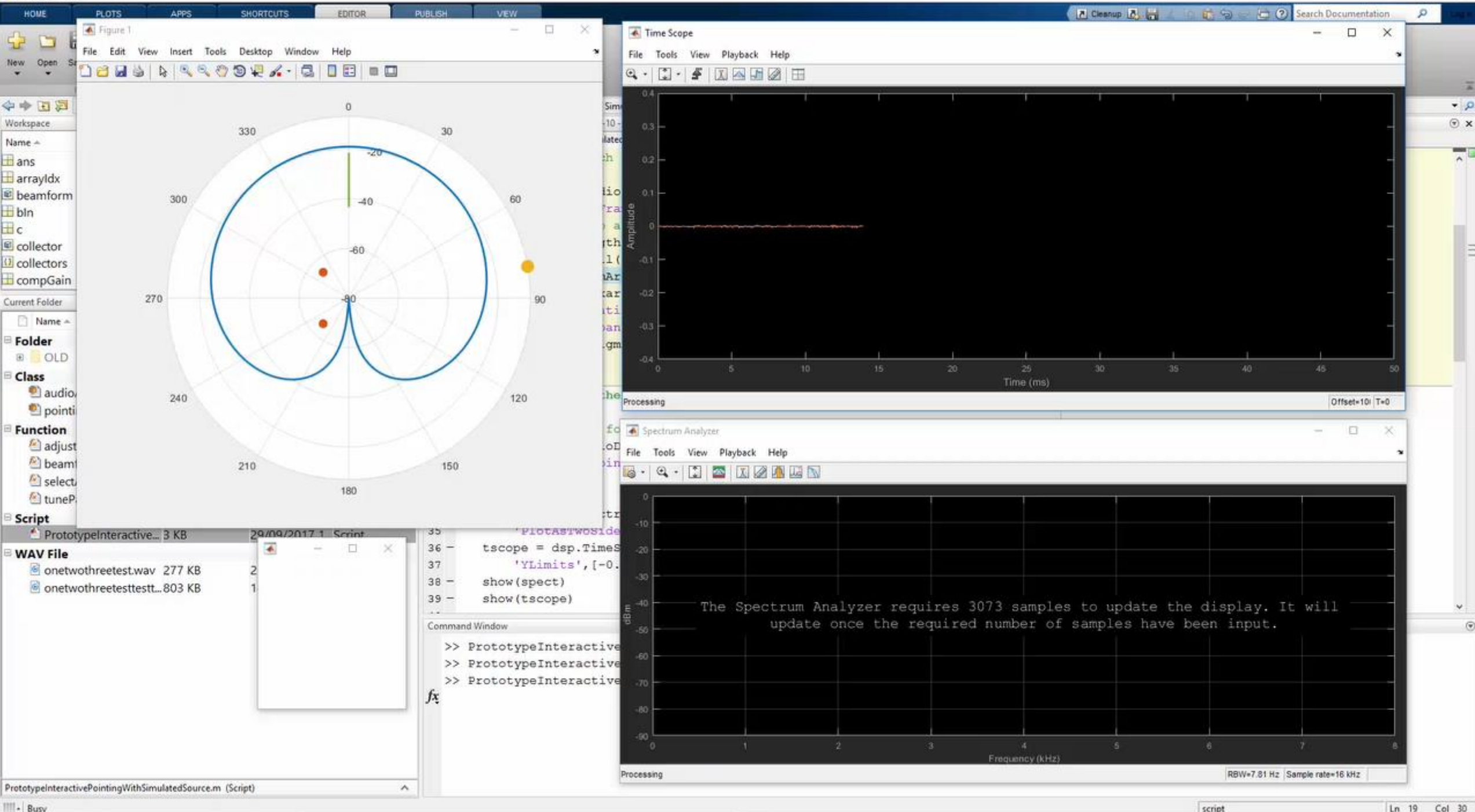
Windows interface showing the MATLAB environment. The top bar includes "LISH" and "VIEW" buttons. The main toolbar contains icons for "Cleanup", "Save", "Undo", "Redo", "Find", "Help", and "Search Documentation". The "Search Documentation" field is active, showing "Log In". Below the toolbar, there are three buttons: "Run and Advance", "Run Section", and "Run and Time". A "RUN" button is also visible.

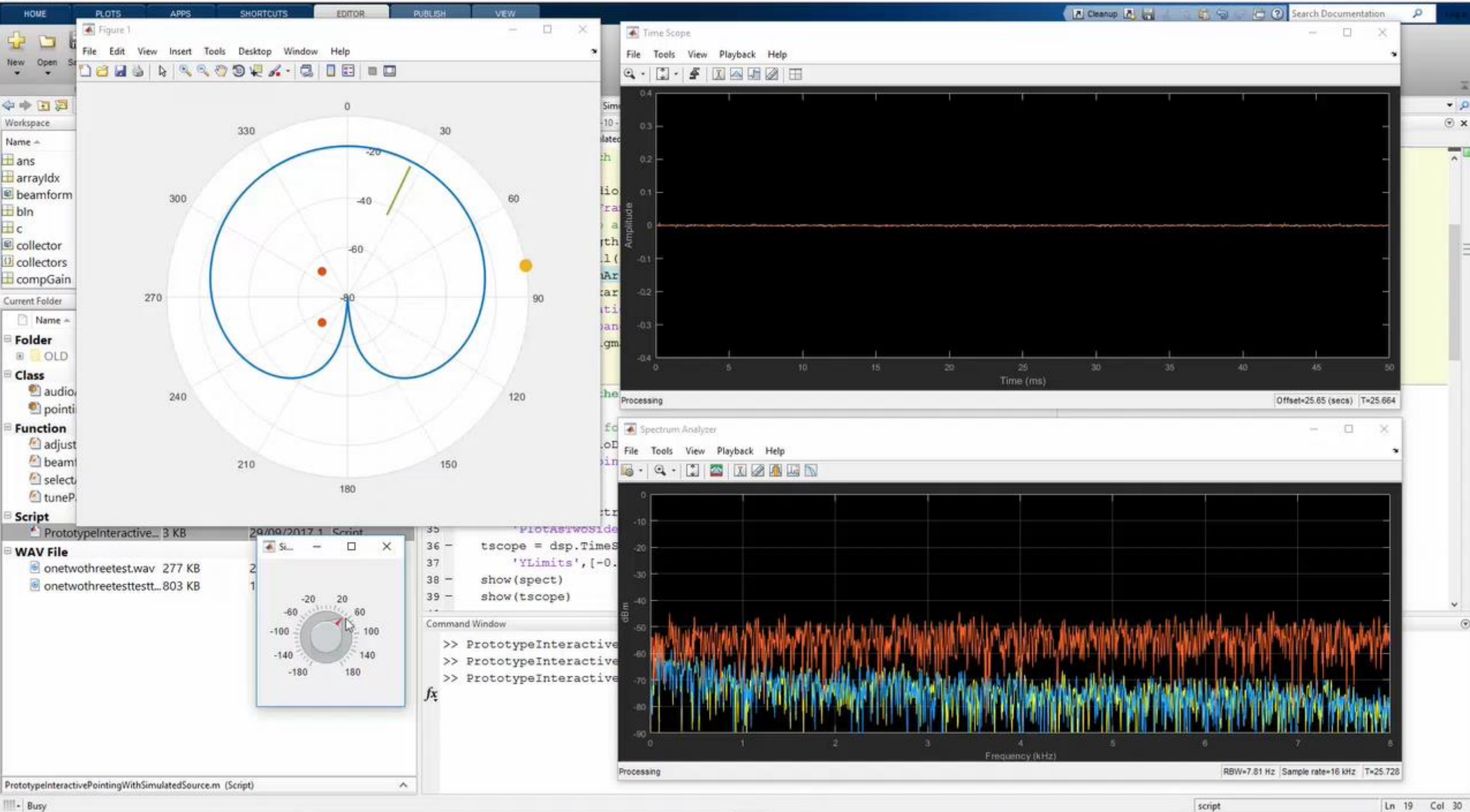
Videos ▶ Code ▶ 4. Array Time Domain Simulation ▶

- C:\Docs\Material\Projects\2017-10 - UK EXPO\Talk PPT\Final Videos\Code\4. Array Time Domain Simulation\SimulateArrayAcquisition.m

SimulateArrayAcquisition.m x PrototypeSystem.m x audioArrayParameters.m x SteerInteractively.m x ValidateArrayDesign.m x adjustPositions.m x +

```
colune = audioArrayParameters(viewer);  
toTune.steeringTheta = -90;  
toTune.sourceTheta = 90;  
  
% Microphone pair  
ula = phased.ULA(2, dx, 'Taper', [1 -1]);  
  
% Acquisition of propagating signal  
collector = phased.WidebandCollector('Sensor',ula,...  
    'PropagationSpeed',c,'SampleRate',fs,'ModulatedInput',false,...  
    'NumSubbands',4096);  
incidentAngle = [toTune.sourceTheta;0];  
  
% Beamformer  
beamform = phased.TimeDelayBeamformer('SensorArray',ula,...  
    'SampleRate',fs,'PropagationSpeed',c,...  
    'DirectionSource','Input port','WeightsOutputPort',false);  
nullSteeringAngle = [toTune.steeringTheta;0];  
  
% Spectrum analysis  
spect = dsp.SpectrumAnalyzer('SampleRate',fs,'PlotAsTwoSidedSpectrum',false,...  
    'YLimits',[-50,15], 'ShowLegend', true,...  
    'SpectralAverages', 128, 'FrequencyScale', 'Log');  
show(spect)  
adjustPositions(viewer, spect)
```

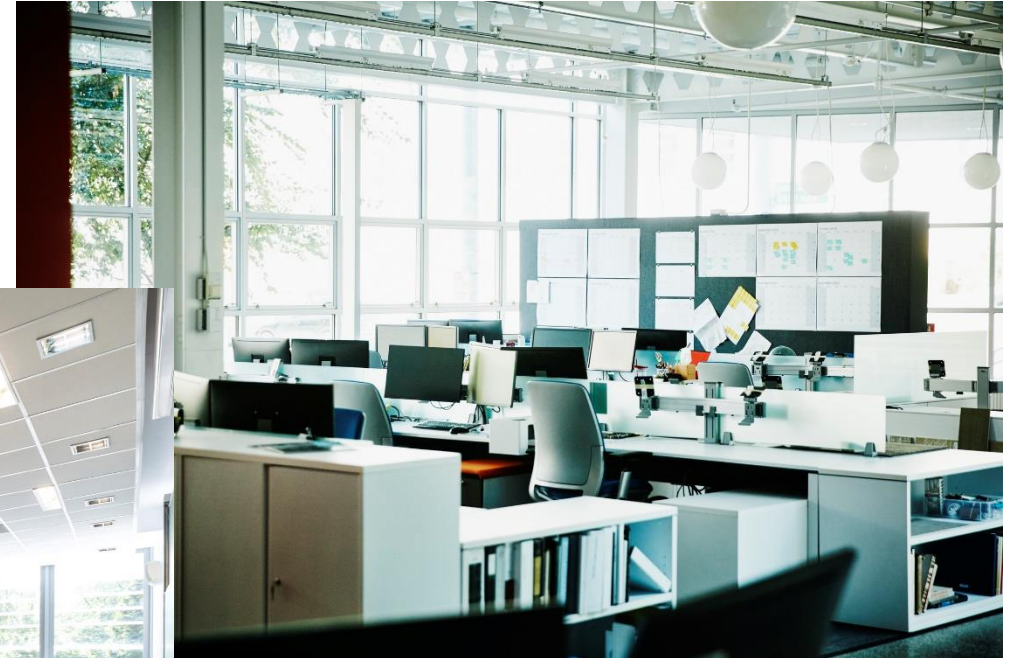






# How Can I ....Validate my voice interface can work in real-life scenarios?

## Constrained Simulations vs. Real Life







PUBLISH

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Run and Time

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nal Videos ▶ Code ▶ 6. Streaming Audio into MATLAB

Editor - C:\Docs\Material\Projects\2017-10 - UK EXPO\Talk PPT\Final Videos\Code\6. Streaming Audio into MATLAB\LiveStreamAndAcquire.m

LiveStreamAndAcquire.m

22 -

release(readaudio)

23

24

%% Stream and visualize live

25

26

% Scope for time-domain visualization

27 -

tscope = dsp.TimeScope('SampleRate',fs,'TimeSpan',0.500,...

28

'YLimits',[-0.5,0.5],'TimeSpanOverrunAction','Scroll');

29

30

% Set shorter block duration

31 -

readaudio.SamplesPerFrame = 1024;

32

33 -

tic

34 -

while toc < 30

35

% Acquire live

36 -

in = readaudio();

37

% Visualize in real-time

38 -

tscope(in);

39 -

end

40

41

%%

42 -

release(readaudio)

Command Window

show [all properties](#)

>> readaudio.Device = 'Microphone (STM32 AUDIO Streaming in FS Mode)'

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FILE

NAVIGATE

EDIT

BREAKPOINTS

RUN

C:\> Docs > Material > Projects > 2017-10 - UK EXPO > Talk PPT > Final Videos > Code > 7. Real-time in MATLAB

Workspace

Editor - C:\Docs\Material\Projects\2017-10 - UK EXPO\Talk PPT\Final Videos\Code\7. Real-time in MATLAB\LevelAndBeamform.m

LevelAndBeamform.m

```
1 classdef LevelAndBeamform < audioPlugin
2     % Copyright 2017 The MathWorks, Inc.
3
4     %#codegen
5     properties
6         % "Looking" direction
7         Theta = 0
8         % Bandpass filter gain (dB) - 0=all bandpassed / -inf=allpass
9         BandpassGain = 0
10        % Beam type
11        Beamtype = BeamformingType.Cardioid
12        % Output Gain (dB)
13        OutputGain = 0
14    end
15
16    properties (Access = private)
17        th = 0
18
19        beamformingType = 3
20    end
```

Command Window

```
fx >>
```

Current Folder

Command History

Name	Size	Date...	Type
<b>Class</b>			
BeamformingType.m	1 KB	11/...	Class
LevelAndBeamform.m	6 KB	13/...	Class
<b>Function</b>			
beamformArraysSpecificationsCG.m	2 KB	11/...	Fun...
selectArrayConfiguration.m	1 KB	14/...	Fun...
<b>WAV File</b>			
AroundThe ArrayAt0_take1.wav	5.6...	28/...	WA...
AroundThe ArrayAt90_take1.wav	7.8...	28/...	WA...

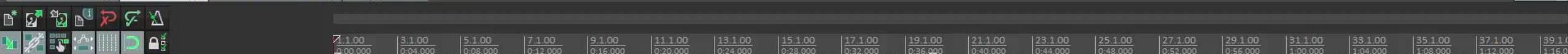
selectArrayConfiguration.m (Function)

Windows Taskbar

Search

23:37

01/10/2017



AroundThe ArrayAt9..1

Input 1..Input 4 [4 c]

monitoring\_record

AroundThe ArrayAt9D\_take1 [chan] 4 [sends] 1 [FX] 3



MASTER

AroundThe ..1

Mixer

The mixer section shows two channels, 1 and 2, with various controls including volume, pan, and mute. Channel 1 has a volume of -16.0 and channel 2 has a volume of -17.4.

1.1.00 / 0:00.000

[Stopped]

BPM 120 4/4 Rate: 1.0

Selection: 1.1.00 20.1.00 19.0.00



FileEditViewInsertItemTrackOptionsActionsHelp

BeatTrackerOnDrums.rpptrackTest.rppEffectsOnDrumsAndGuitar.rppNoisySpeech.RPP

1.001.302.102.303.103.304.104.305.105.306.106.307.107.308.108.309.109.3010.110.311.111.312.112.313.113.314.114.315.115.3

NoisySpeech-16-22p5-mono-5secs.wav

NoisySpeech-16-22p5-

IN

Input 1

1

VST: ERA-N (accusonus) - Track 1 "NoisySpeech-16-22p5-mono-5secs.wav"

No preset+Param2 in 2 outUI

ERA-Naccusonus

100

0dB

LEVEL

BYPASS

PROCESSING40%

A

Standard license

Select time

1.1.00 / 0:00.000

[Stopped]

BPM1204/4Rate:1.0

Selection:1.1.001.1.000.0.00

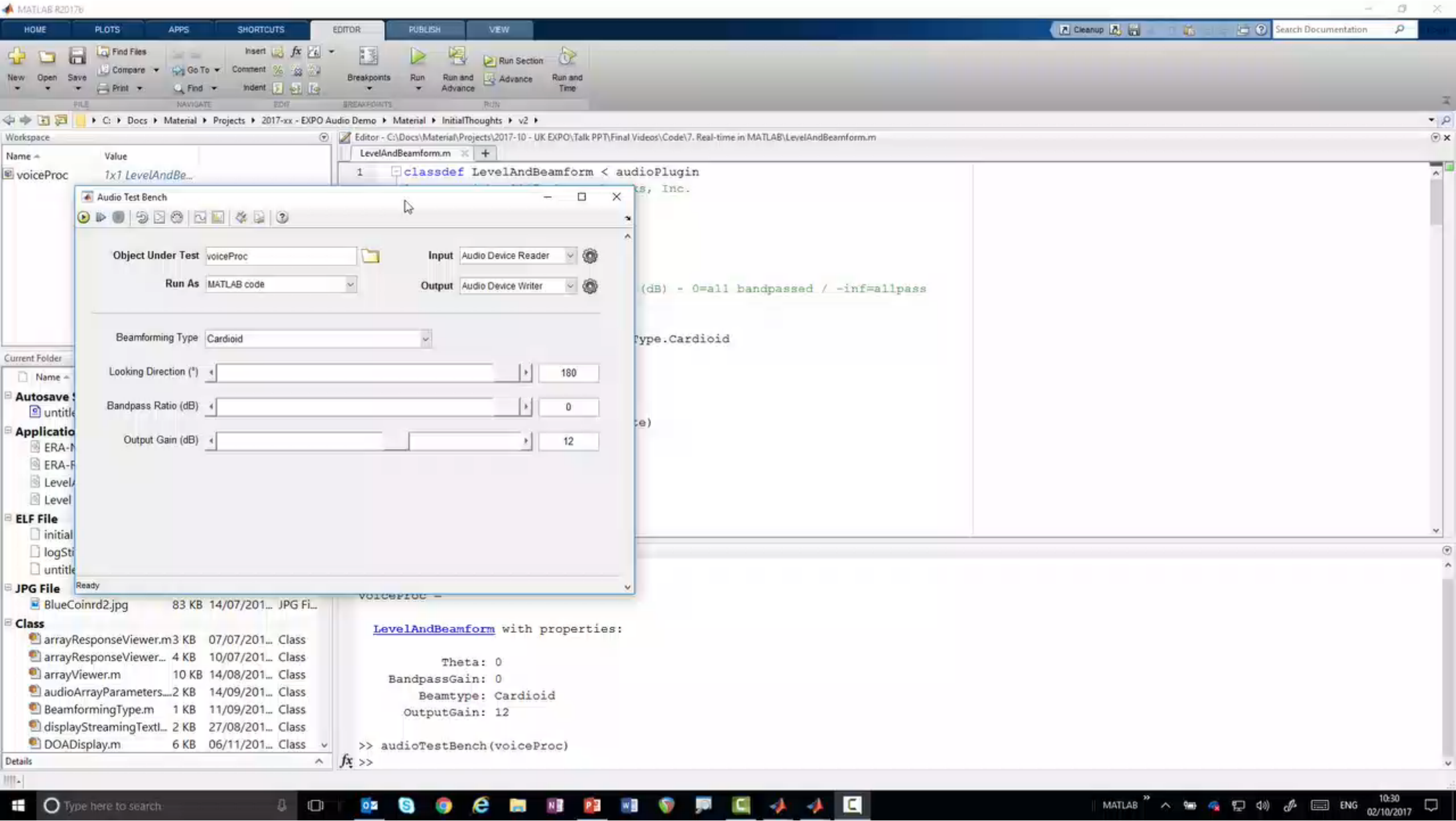
MASTERNoisySpeech-1

0.31286-1812-1824-30-36-42-54-58.5

MMS

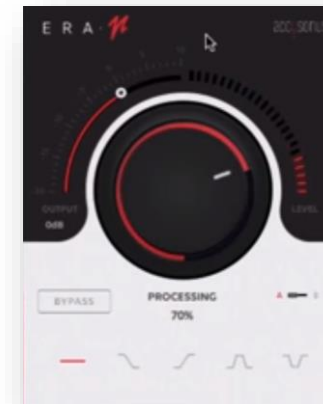
1





# How can I...

2. Validate my voice interface can work in real-life scenarios?
3. Understand what else can help me improve my performance?





## Plugin hosting

```
>> noiseRemover = loadAudioPlugin('ERA-N.vst')
```

```
noiseRemover =
```

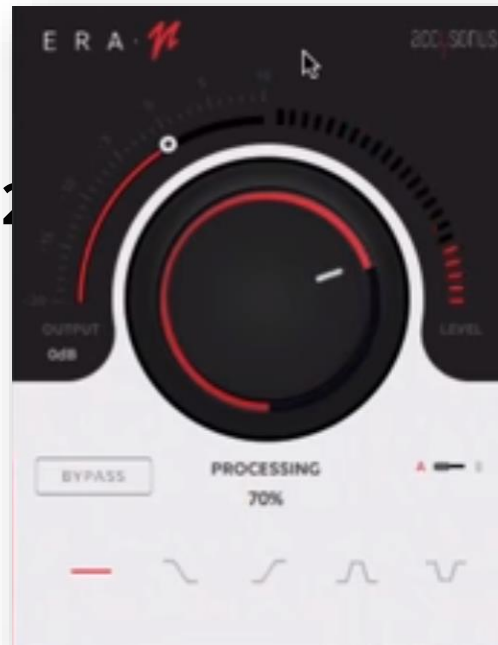
```
VST plugin 'ERA-N' 2 in, 2
```

```
Processing: 40 %
```

```
Gain: 0 dB
```

```
Tilt: 'NoTilt'
```

```
Bypass: 0
```



<https://accusonus.com/products/era-n>

```
>> noiseRemover.Processing = 60;
```

```
>> noiseRemover.Gain = 3;
```

```
>> y = process(noiseRemover, x)
```



# How To Measure Performance?

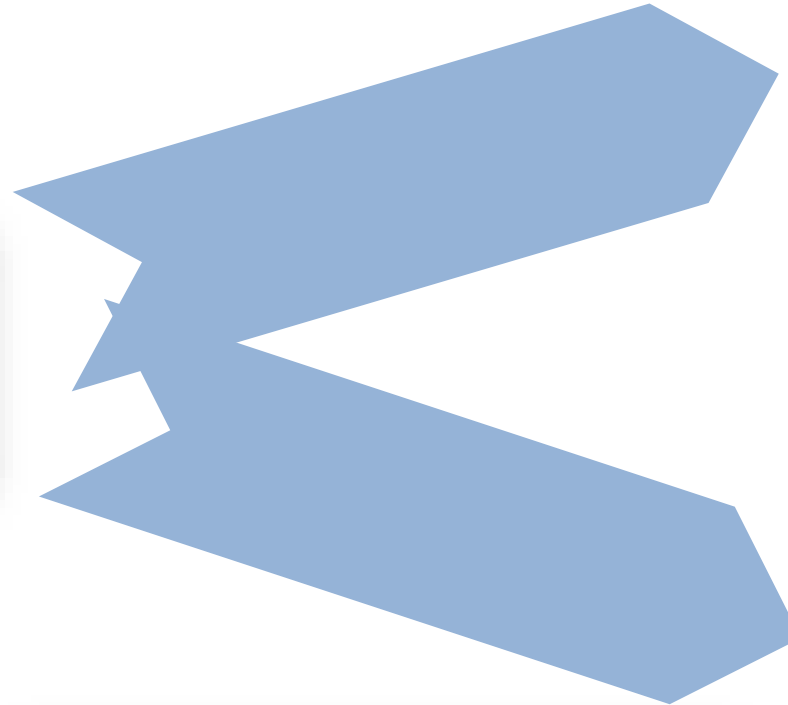
```

while ishandle(hui)

    % Audio acquisition (from file)
    rxAll = readaudio();
    inAll = levelChannels(rxAll);

    % Event segmentation and DOA
    [toTune.doa, segmentation, eventsDetected] = bufferAndSearch(inAll, fs, searchBufferLength);
    % Point array towards estimated DOA
    toTune.pointingTheta = toTune.doa;

    % Select correct subarray for pointing
  
```



Output audio  
"sounds good"



True	Guessed
e slid on the smooth planks"	"the birth canal we sl
in stockings is hard to sell"	"a large size in stock
to the dark blue background"	"blue the sheet to the
ll the depth of a well"	"it's easy to tell the
hicken leg is a rare dish"	"these days a chicken
served in round bowls"	"rice is often served

"91.5% of spoken  
sentences correctly  
converted"

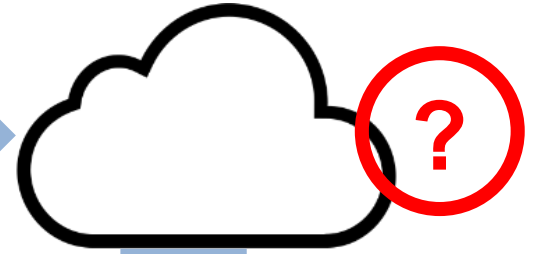
```

while ishandle(hui)

    % Audio acquisition (from file)
    rxAll = readaudio();
    inAll = levelChannels(rxAll);

    % Event segmentation and DOA
    [toTune.doa, segmentation, eventsDetected] = bufferAndSearch(inAll, fs, searchBufferLength);
    % Point array towards estimated DOA
    toTune.pointingTheta = toTune.doa;

    % Select correct subarray for pointing
    % Select correct subarray for pointing
    
```



Current Folder

Name	Size
Folder	
Harvard H1	
H1 Gab BlueCoin...	
1.wav	231 KB
2.wav	217 KB
3.wav	209 KB
4.wav	249 KB
5.wav	269 KB
6.wav	251 KB
7.wav	253 KB
8.wav	269 KB
9.wav	277 KB
10.wav	269 KB
H1 Gab BlueCoin...	
1.wav	241 KB
2.wav	227 KB
3.wav	239 KB
4.wav	293 KB
5.wav	269 KB
6.wav	253 KB
7.wav	321 KB
8.wav	245 KB
9.wav	311 KB
10.wav	287 KB
H1 Gab Headset	
1.wav	253 KB
2.wav	263 KB



True	GuessedBaseline	ConfidenceBaseline
"the birch canoe slid on the smooth planks"	"the birth canal we slid on the smooth planks"	0.74933
" a large size in stockings is hard to sell"	"a large size in stockings is hard to sell"	0.9366
"glue the sheet to the dark blue background"	"blue the sheet to the dark blue background"	0.95009
"its easy to tell the depth of a well"	"it's easy to tell the depth of a well"	0.97403
"these days a chicken leg is a rare dish"	"these days a chicken leg is a redfish"	0.86686
"rice is often served in round bowls"	"rice is often served in rum balls"	0.95946
"the juice of lemons makes fine punch"	"the cheese of lemons makes flying punch"	0.79089
"the box was thrown beside the parked truck"	"the box was thrown beside the box truck"	0.82796
"the hogs were fed chopped corn and garbage"	"the Hawks were fed chops corn and garbage"	0.81863
"four hours of steady work faced us"	"4 hours of study world fastest"	0.82847

New Tab



# Test performance with speech-to-text services

```
>> [samples, fs] = audioread('helloaudioPD.wav');  
>> soundsc(samples, fs)  
>> speechObject = speechClient('Google', 'languageCode', 'en-US');  
>> outInfo = speech2text(speechObject, samples, fs);
```

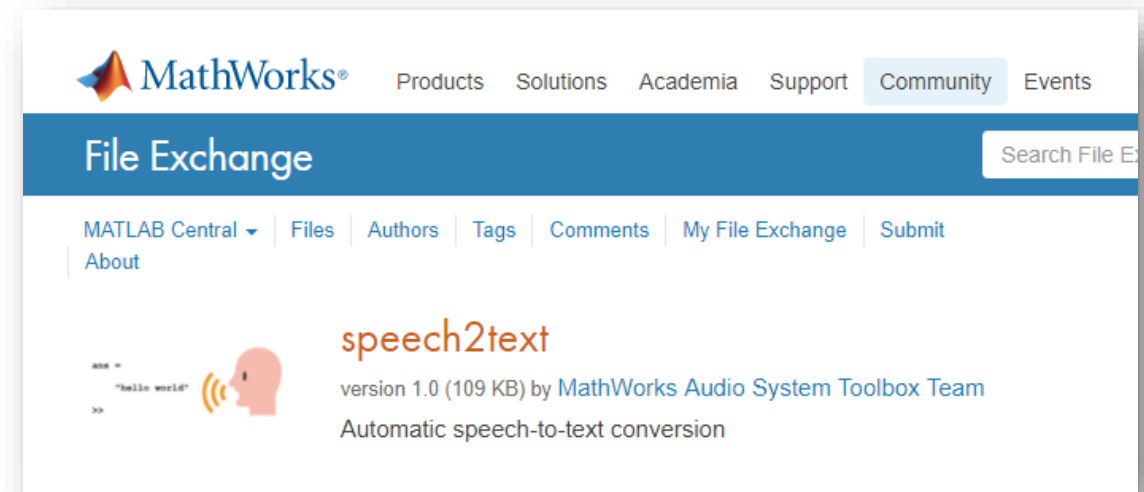


```
>> outInfo.TRANSCRIPT =
```

```
ans =  
    'hello audio product Developers'
```

```
>> outInfo.CONFIDENCE =
```

```
ans =  
    0.9385
```



<https://www.mathworks.com/matlabcentral/fileexchange/65266-speech2text>



```

while ishandle(hui)

    % Audio acquisition (from file)
    rxAll = readaudio();
    inAll = levelChannels(rxAll);

    % Event segmentation and DOA
    [toTune.doa, segmentation, eventsDetected] = bufferAndSearch(inAll, fs, searchBufferLength);
    % Point array towards estimated DOA
    toTune.pointingTheta = toTune.doa;

    % Select correct subarray for pointing
    % Select correct subarray for pointing
    
```



Current Folder

Folder	Name	Size
Harvard H1		
H1 Gab BlueCoin...		
	1.wav	231 KB
	2.wav	217 KB
	3.wav	209 KB
	4.wav	249 KB
	5.wav	269 KB
	6.wav	251 KB
	7.wav	253 KB
	8.wav	269 KB
	9.wav	277 KB
	10.wav	269 KB
H1 Gab BlueCoin...		
	1.wav	241 KB
	2.wav	227 KB
	3.wav	239 KB
	4.wav	293 KB
	5.wav	269 KB
	6.wav	253 KB
	7.wav	321 KB
	8.wav	245 KB
	9.wav	311 KB
	10.wav	287 KB
H1 Gab Headset		
	1.wav	253 KB
	2.wav	263 KB



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"the box was thrown beside the parked truck"	"the box was thrown beside the box truck"	0.82796
"the hogs were fed chopped corn and garbage"	"the Hawks were fed chops corn and garbage"	0.81863
"four hours of steady work faced us"	"4 hours of study world fastest"	0.82847

# Building a small speech dataset quickly

**Dataset Recorder**

Rec Play Store

Recording Subfolder: Gabriele

SpeechContent: one two three test test test (x2)

SpeakerName: Gabriele Bunkheila

SpeakerGender: Male

SpeakerAge: 41

Language: English

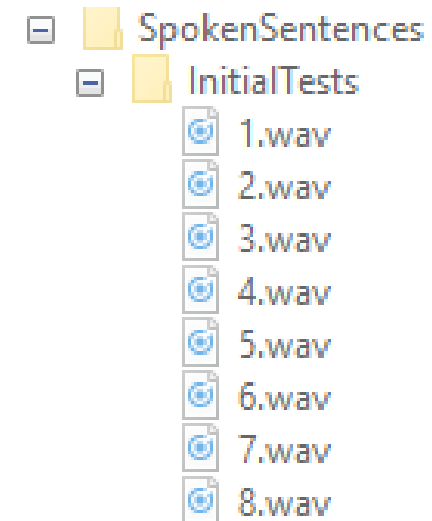
Device: Plantronics Blackwire 325

MicrophoneDistance: Close

EnvironmentType: Home

EnvironmentSize: Small

Sentiment: Neutral



Sentiment	Filename	SampleRate	NumChannels	NumSamples	DateRecorded
"Neutral"	"Neutral\1.wav"	16000	2	73728	11-Sep-2017 11:59:57
"Neutral"	"Neutral\2.wav"	16000	2	62976	11-Sep-2017 12:00:22
"Neutral"	"Neutral\3.wav"	16000	2	1.1674e+05	11-Sep-2017 12:01:00
"Neutral"	"Neutral\4.wav"	16000	2	1.0701e+05	11-Sep-2017 12:01:22
"Neutral"	"Neutral\5.wav"	16000	2	60416	11-Sep-2017 12:01:41
"Neutral"	"Neutral\6.wav"	16000	2	1.9251e+05	11-Sep-2017 12:02:30
"Neutral"	"Neutral\7.wav"	16000	2	1.3107e+05	11-Sep-2017 12:03:04
"Neutral"	"Neutral\8.wav"	16000	2	64512	11-Sep-2017 12:03:23

# Building a small speech dataset quickly

## How to accelerate speech content labelling?

### H1 Harvard Sentences

1. The birch canoe slid on the smooth planks.
2. Glue the sheet to the dark blue background.
3. It's easy to tell the depth of a well.
4. These days a chicken leg is a rare dish.
5. Rice is often served in round bowls.
6. The juice of lemons makes fine punch.
7. The box was thrown beside the parked truck.
8. The hogs were fed chopped corn and garbage.
9. Four hours of steady work faced us.
10. A large size in stockings is hard to sell.

### H2 Harvard Sentences

1. The boy was there when the sun rose.
2. A rod is used to catch pink salmon.
3. The source of the huge river is the clear spring.
4. Kick the ball straight and follow through.
5. Help the woman get back to her feet.
6. A pot of tea helps to pass the evening.
7. Smoky fires lack flame and heat.
8. The soft cushion broke the man's fall.
9. The salt breeze came across from the sea.
10. The girl at the booth sold fifty bonds.

### H3 Harvard Sentences

1. The small pup gnawed a hole in the sock.
2. The fish twisted and turned on the bent hook.
3. Press the pants and sew a button on the vest.
4. The swan dive was far short of perfect.
5. The beauty of the view stunned the young boy.
6. Two blue fish swam in the tank.
7. Her purse was full of useless trash.
8. The colt reared and threw the tall rider.
9. It snowed, rained, and hailed the same morning.
10. Read verse out loud for pleasure.

### H4 Harvard Sentences

1. Hoist the load to your left shoulder.
2. Take the winding path to reach the lake.
3. Note closely the size of the gas tank.
4. Wipe the grease off his dirty face.
5. Mend the coat before you go out.
6. The wrist was badly strained and hung limp.

```
>> harvard{1}
```

```
ans =
```

```
10×1 string array
```

```
"the birch canoe slid on the smooth planks"  
"glue the sheet to the dark blue background"  
"its easy to tell the depth of a well"  
"these days a chicken leg is a rare dish"  
"rice is often served in round bowls"  
"the juice of lemons makes fine punch"  
"the box was thrown beside the parked truck"  
"the hogs were fed chopped corn and garbage"  
"four hours of steady work faced us"  
" a large size in stockings is hard to sell"
```

<http://www.cs.cmu.edu/afs/cs.cmu.edu/project/fgdata/OldFiles/Recorder.app/utterances/Type1/harvsents.txt>

# Building a small speech dataset quickly

## Example: an App with automated content labelling

**Dataset Recorder**

Rec Play Store

Recording Subfolder: H1 Gab BlueCoin Close

SpeechContent: the birch canoe slid on the smooth planks

SpeakerName: the birch canoe slid on the smooth planks  
glue the sheet to the dark blue background  
its easy to tell the depth of a well  
these days a chicken leg is a rare dish  
rice is often served in round bowls  
the juice of lemons makes fine punch  
the box was thrown beside the parked truck  
the hogs were fed chopped corn and garbage  
four hours of steady work faced us  
a large size in stockings is hard to sell

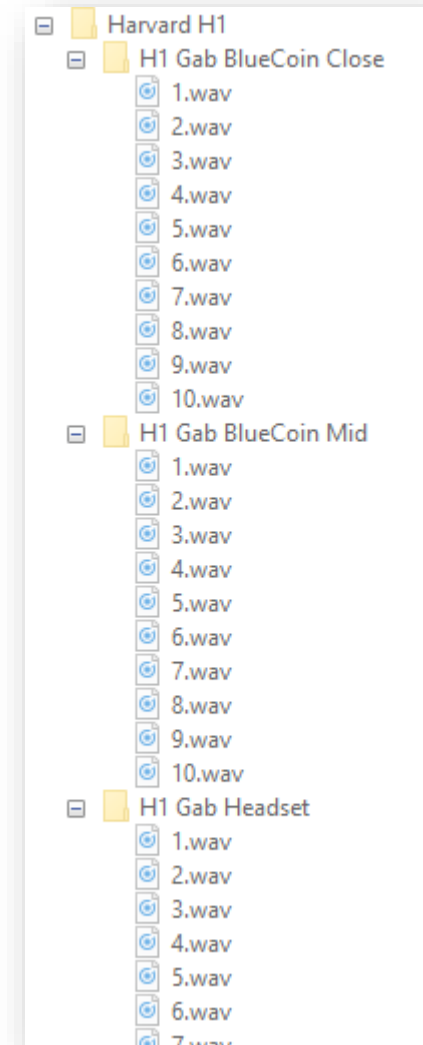
SpeakerGender:

SpeakerAge:

Language:

Device: BlueCoin beta

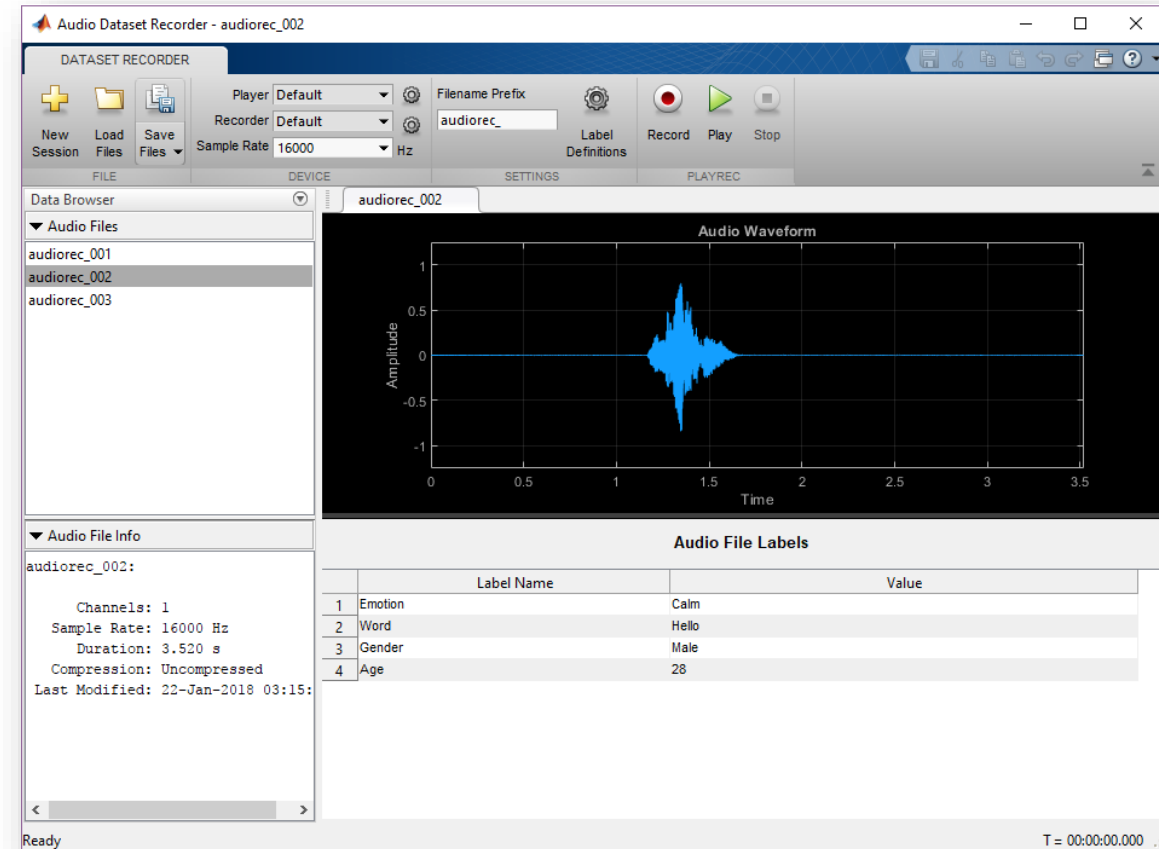
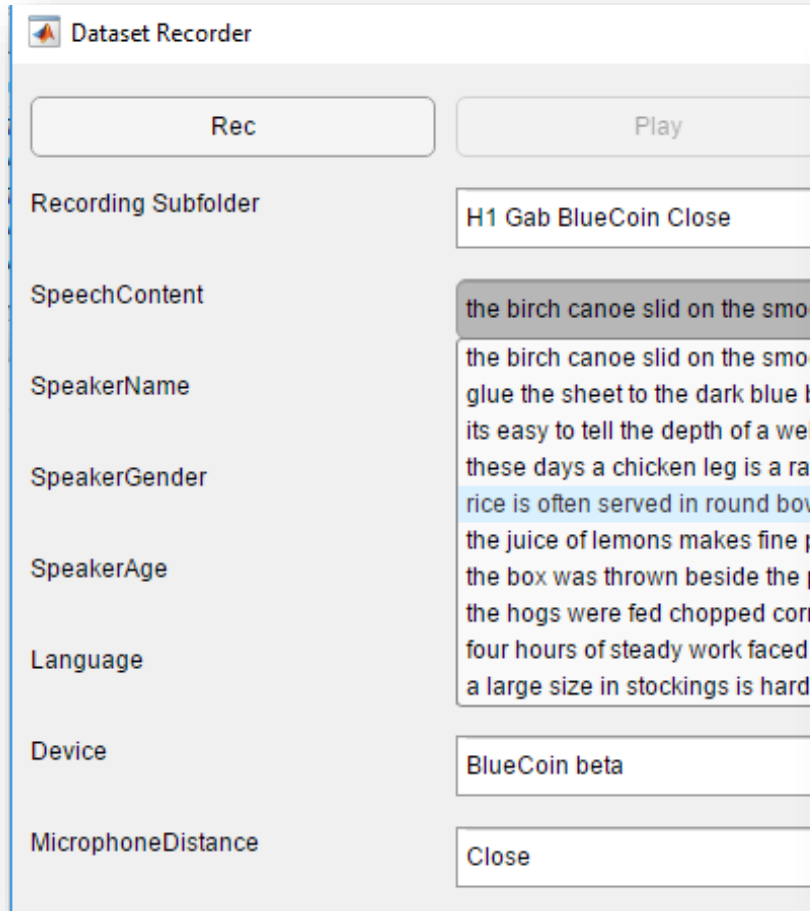
MicrophoneDistance: Close





# Building a small speech dataset quickly

## Example: an App with automated content labelling\*



**\*See also Dataset Recorder App prototype in example "Record Audio Datasets" (From R2018a in Audio System Toolbox)**

```

while ishandle(hui)

    % Audio acquisition (from file)
    rxAll = readaudio();
    inAll = levelChannels(rxAll);

    % Event segmentation and DOA
    [toTune.doa, segmentation, eventsDetected] = bufferAndSearch(inAll, fs, searchBufferLength);
    % Point array towards estimated DOA
    toTune.pointingTheta = toTune.doa;

    % Select correct subarray for pointing
    % Select correct subarray for pointing
    
```



Current Folder		
Name	Size	
Folder		
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1.wav	231 KB	
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9.wav	277 KB	
10.wav	269 KB	
H1 Gab BlueCoin...	277 KB	253 KB
1.wav	241 KB	269 KB
2.wav	227 KB	277 KB
3.wav	239 KB	241 KB
4.wav	293 KB	227 KB
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10.wav	287 KB	245 KB
H1 Gab Headset	311 KB	321 KB
1.wav	253 KB	287 KB
2.wav	263 KB	311 KB



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Workspace Command History Current Folder Editor - assessSpeechRecognitionPerformance.m Command Window

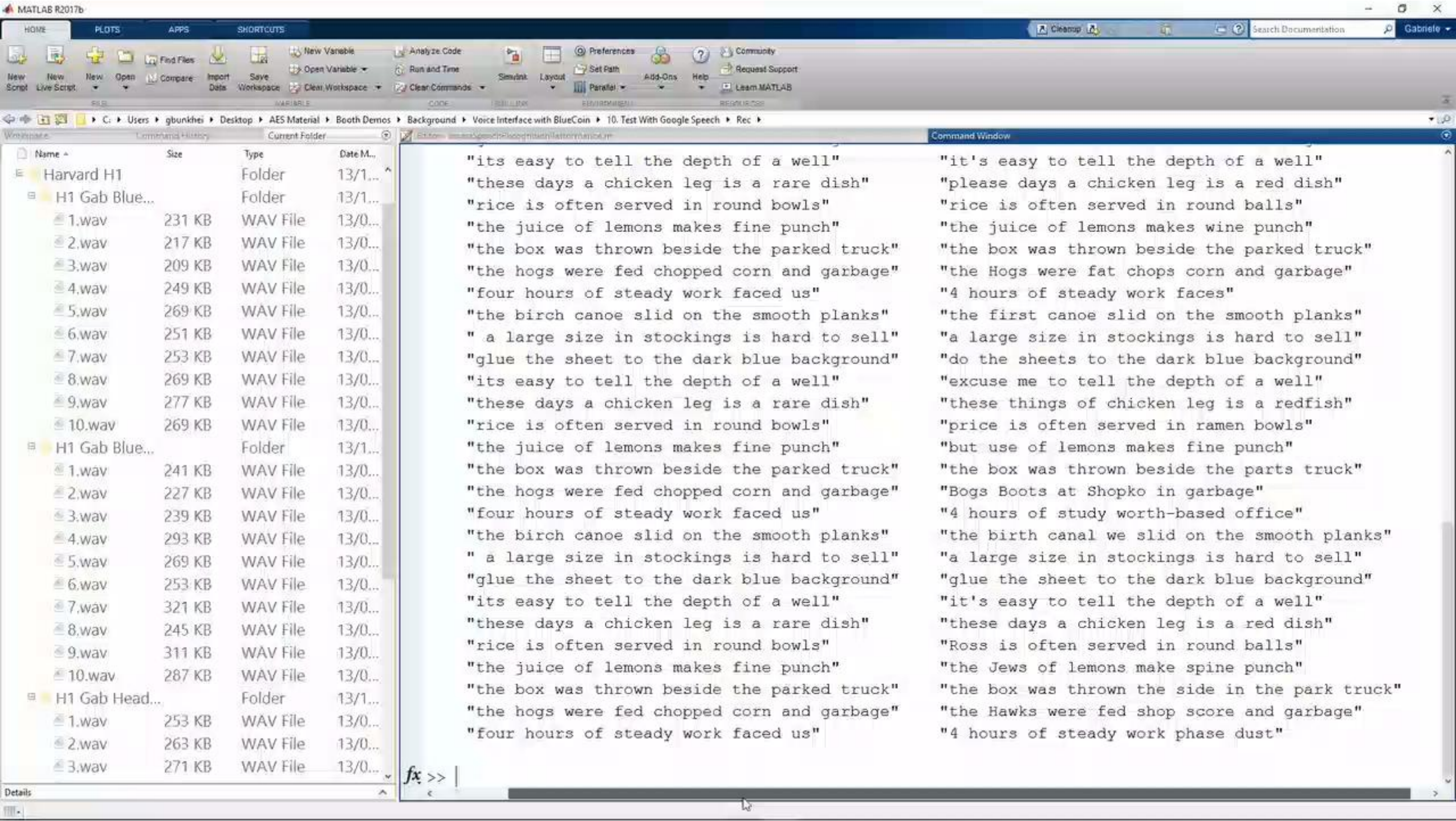
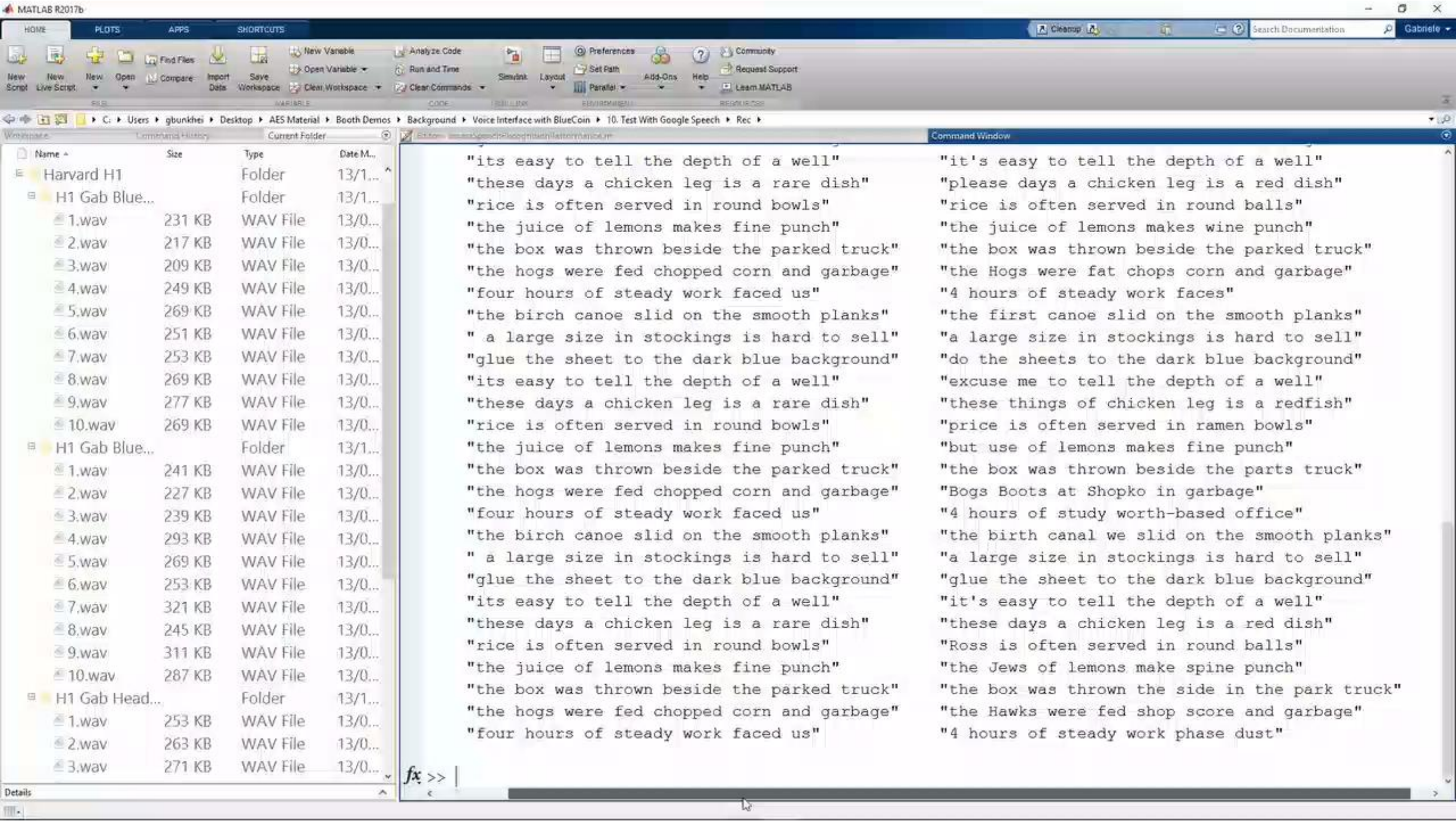
Name	Size	Type	Date M...
Harvard H1		Folder	13/1...
H1 Gab BlueC...		Folder	13/1...
1.wav	231 KB	WAV File	13/0...
2.wav	217 KB	WAV File	13/0...
3.wav	209 KB	WAV File	13/0...
4.wav	249 KB	WAV File	13/0...
5.wav	269 KB	WAV File	13/0...
6.wav	251 KB	WAV File	13/0...
7.wav	253 KB	WAV File	13/0...
8.wav	269 KB	WAV File	13/0...
9.wav	277 KB	WAV File	13/0...
10.wav	269 KB	WAV File	13/0...
H1 Gab BlueC...		Folder	13/1...
1.wav	241 KB	WAV File	13/0...
2.wav	227 KB	WAV File	13/0...
3.wav	239 KB	WAV File	13/0...
4.wav	293 KB	WAV File	13/0...
5.wav	269 KB	WAV File	13/0...
6.wav	253 KB	WAV File	13/0...
7.wav	321 KB	WAV File	13/0...
8.wav	245 KB	WAV File	13/0...
9.wav	311 KB	WAV File	13/0...
10.wav	287 KB	WAV File	13/0...
H1 Gab Head...		Folder	13/1...
1.wav	253 KB	WAV File	13/0...
2.wav	263 KB	WAV File	13/0...
3.wav	271 KB	WAV File	13/0...

```

15  %% Cycle through recordings
16  % Init arrays to store results
17  numFiles = length(ds.Files);
18  k = 0;
19
20  conf = zeros(1,numFiles);
21  trueStrings = repmat("",1,numFiles);
22  guessedStrings = repmat("",1,numFiles);
23
24  while hasdata(ds)
25      % Read audio file content
26      entry = read(ds);
27      x = entry.samples;
28
29      fprintf('Sending "%s"...\\n',entry.labels.SpeechContent)
30
31      % Send ORIGINAL audio to cloud service and get transcription
32      [outString, outInfo] = speech2text(x(:,1) , entry.fs);
33
34      % Store results
35      k = k + 1;
36      conf(k) = outInfo.confidence;
37      trueStrings(k) = entry.labels.SpeechContent;
38      guessedStrings(k) = outString;
39
40  end
41
42  tResults = table(trueStrings', guessedStrings', num2str(conf'),...
43      'VariableNames',{'True','Guessed','Confidence'},...

```

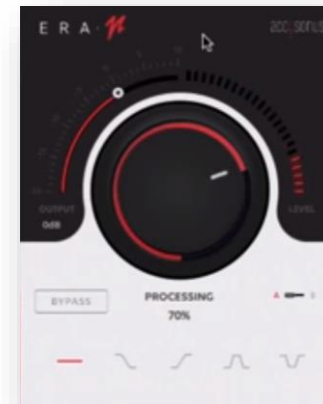
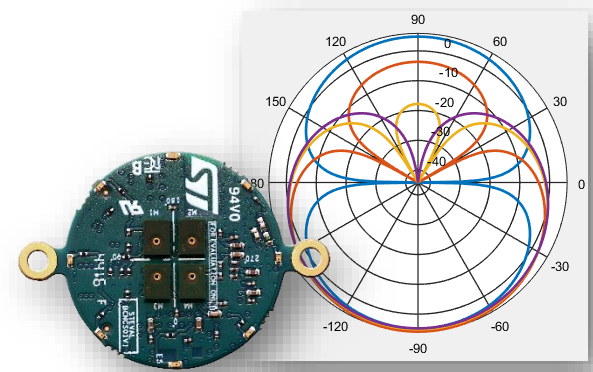






# How Can I...

1. Design a microphone array system?
2. Validate my voice interface can work in real-life scenarios?
3. Understand what else can help me improve my performance?





# Summary

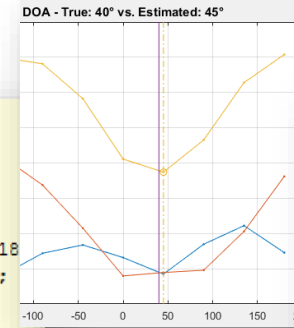
- Innovate
- Reuse
- Prototype

```
% Direction to estimate
incidentAngle = [40;0];

% Theta increments
thetaStep = 45;
steeringThetas = -180:thetaStep:180;
steeringAngles = [steeringThetas;

frameSize = 512;

s = beamformArraysSpecificationsDOA(
```



Dataset Recorder

Rec Play

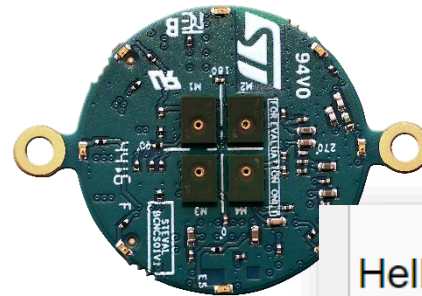
Recording Subfolder: H1 Gab BlueCoin Close

SpeechContent: the birch canoe slid on the

SpeakerName: Gabriele Bunkheila

SpeakerGender: Male

SpeakerAge: 41



Hello  
Anyone out there

