

### Voice Recognition through Machine Learing



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# 4 Transfering the weights to Raspberry Pi (Yet to be done)

Abstract—This manual shows how to develop a voice recognition algorithm and use it to control a toycar.

#### 1 Dataset

1.1 Draw the block diagram of the AI-ML system for the toycar.

**Solution:** See Fig. 1.1

- 1.2 Record 'forward' 80 times using you phone and save as 'forwardi.wav' for i = 1, ..., 80. The recording duration should be between 1-3 seconds.
- 1.3 Repeat by recording 'left', 'right', 'back' and 'stop'. Make sure that the audio files for each command are in separate directories. Download the following directory for reference

svn checkout https://github.com/gadepall/ EE1390/trunk/AI-ML/audio\_dataset

1.4 Use the following script to generate a dataset for 'back' command. Explain through a block diagram.

https://raw.githubusercontent.com/gadepall/ EE1390/master/AI-ML/codes/250files.py

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**Solution:** The datasets are generated through zero padding. The diagram in Fig. 1.4 explains how this is done for the back command.

- 1.5 Suitably modify the above script to generate similar datasets for 'left', 'right', 'stop' and 'forward'.
- 1.6 Summarize the datasets generated through a table.

**Solution:** See Table 1.6

#### 2 Linear Regression: Least Souares

- 2.1 Draw the block diagram for the ML algorithm **Solution:** See Fig. 2.1
- 2.2 List the reference vectors for all the voice commands.

**Solution:** See Table 2.2.

2.3 The sigmoid function is defined as

$$s(x) = \frac{1}{1 + e^{-x}} \tag{2.1}$$

Sketch s(x).

**Solution:** The following code plots s(x) in Fig. 2.3.

2.4 Show that 0 < s(x) < 1.

**Solution:** s(x) is useful for transforming large values to a value between 0 and 1.

2.5 Formulate a regression model for the voice recognition system.

**Solution:** Let  $\mathbf{x}$  be the voice command. The model used is

$$\mathbf{v} = \mathbf{W}\mathbf{x} + \mathbf{b} \tag{2.2}$$

where **W** is the weight matrix and **b** is the bias vector. Ideally, the output **y** should be one of the reference vectors in Table 2.2.

2.6 Frame an optimization problem for estimating W and b.

Solution: This is done by considering the cost

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Commands	Commands Input		Conditioned	Training	Testing
		Input file			
Back	80	250	20000	16000	4000
Forward	80	250	20000	16000	4000
Left	80	250	20000	16000	4000
Right	80	250	20000	16000	4000
Stop	80	250	20000	16000	4000
	Total		100000		

TABLE 1.6: File calculus

Command	Reference vector						
Forward	(1	0	0	0	$0)^T$		
Back	(0	1	0	0	$0)^T$		
Left	(0	0	1	0	$0)^T$		
Right	(0	0	0	1	$0)^T$		
Stop	(0	0	0	0	$1)^T$		

TABLE 2.2: Reference vectors

function

$$\min_{\mathbf{W}, \mathbf{b}} J(\mathbf{W}, \mathbf{b}) = \frac{1}{2} ||\mathbf{y} - \hat{\mathbf{y}}||^2$$

where

$$\hat{\mathbf{y}} = \mathbf{W}\mathbf{x} + \mathbf{b} \tag{2.4}$$

3 Gradient Descent

3.1 Show that

$$s'(x) = (1 - s(x)) s(x)$$
 (3.1)

3.2 Let **W** be  $2 \times 2$  and **x**, **y** be  $2 \times 1$ . Show that

$$\frac{\partial \left( \mathbf{y}^T \mathbf{W} \mathbf{x} \right)}{\partial \mathbf{W}} = \mathbf{y} \mathbf{x}^T \tag{3.2}$$

3.3 Given that **b** is  $2 \times 1$ , show that

$$\frac{\partial ||\mathbf{W}\mathbf{x} + \mathbf{b}||^2}{\partial \mathbf{W}} = 2(\mathbf{W}\mathbf{x} + \mathbf{b})\mathbf{x}^T$$

3.4 Find

$$\frac{\partial ||\mathbf{y} - \hat{\mathbf{y}}||^2}{\partial \mathbf{W}} \tag{3.4}$$

**Solution:** 

$$\|\mathbf{y} - \hat{\mathbf{y}}\|^2 = \|y\|^2 - 2\mathbf{y}^T (\mathbf{W}\mathbf{x} + \mathbf{b}) + \|\mathbf{W}\mathbf{x} + \mathbf{b}\|^2$$
(3.5)

From (3.2) and (3.3)

$$\frac{\partial ||\mathbf{y} - \hat{\mathbf{y}}||^2}{\partial \mathbf{W}} = -2(\mathbf{y} - \hat{\mathbf{y}})\mathbf{x}^T$$
 (3.6)

3.5 Show that

$$\frac{\partial \mathbf{y}^T \mathbf{b}}{\partial \mathbf{b}} = \mathbf{y} \tag{3.7}$$

3.6 Show that

$$\frac{\partial ||\mathbf{b}||^2}{\partial \mathbf{b}} = 2\mathbf{b} \tag{3.8}$$

3.7 Show that

(2.3)

$$\frac{\partial \|\mathbf{W}\mathbf{x} + \mathbf{b}\|^2}{\partial \mathbf{b}} = 2(\mathbf{W}\mathbf{x} + \mathbf{b})$$
 (3.9)

3.8 Show that

$$\frac{\partial ||\mathbf{y} - \hat{\mathbf{y}}||^2}{\partial \mathbf{h}} = -2(\mathbf{y} - \hat{\mathbf{y}}) \tag{3.10}$$

3.9 W and b can be estimated from (2.3) using

$$\mathbf{W}(n+1) = \mathbf{W}(n) - \frac{\alpha}{2} \frac{\partial J(\mathbf{W}, \mathbf{b})}{\partial \mathbf{W}}$$
(3.11)

$$\mathbf{b}(n+1) = \mathbf{b}(n) - \frac{\alpha}{2} \frac{\partial J(\mathbf{W}, \mathbf{b})}{\partial \mathbf{b}}$$
(3.12)

Show that (3.11) can be expressed as

$$\mathbf{W}(n+1) = \mathbf{W}(n) + \alpha \left(\mathbf{y} - \hat{\mathbf{y}}\right) \mathbf{x}^{T}$$
 (3.13)

$$\mathbf{b}(n+1) = \mathbf{b}(n) + \alpha (\mathbf{y} - \hat{\mathbf{y}}) \tag{3.14}$$

3) 3.10 Store the complete dataset in a directory and

run **code.py** from within the directory. Note that this should be done on a powerful workstation. This will generate two files **W1.out** and **b.out**.

#### 3.1 Python code

https://github.com/raktimgg/ML-algorithm-for-speech-recording is the full code that is used for training. The accuracy we are getting is around 98 percent.

#### 3.2 Dataset

We have made our own dataset by recording 25 samples of each word. Each of these samples are recreated by adding empty elements in the front and back in many different cobinations to create a dataset of 6250 samples for each word. All the audio files are imported to an array in the code and converted to mfcc format before training. For creating training dataset we recorded 25 audio file of each of the following word -

- 1)Forward
- 2)Left
- 3)Right
- 4)Back
- 5)Stop

The code for generating 6250 samples for each word from 25 samples can be found in the github link attached.

https://github.com/abhishekbairagi/Making-Dataset-for-ML/t

## 4 Transfering the weights to Raspberry Pi (Yet to be done)

The weight(W1 and B) are saved in a file at the end of the code. These weights will be transferred to the raspberry pi and a simple program written, will record audio on the raspberry pi, do the calculations using the weights and predict the text output. This output will be sent, using bluetooth, to the toy car, which will move accordingly.

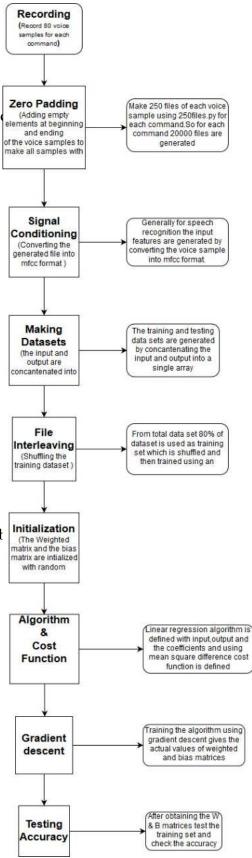


Fig. 1.1: ML System

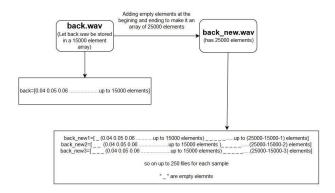


Fig. 1.4: Zero padding

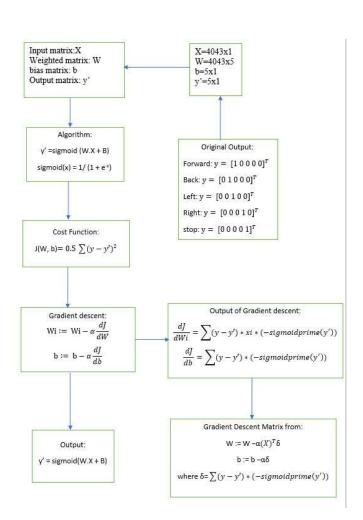


Fig. 2.1: Least squares and gradient descent

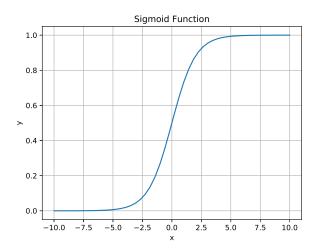


Fig. 2.3: Sigmoid function