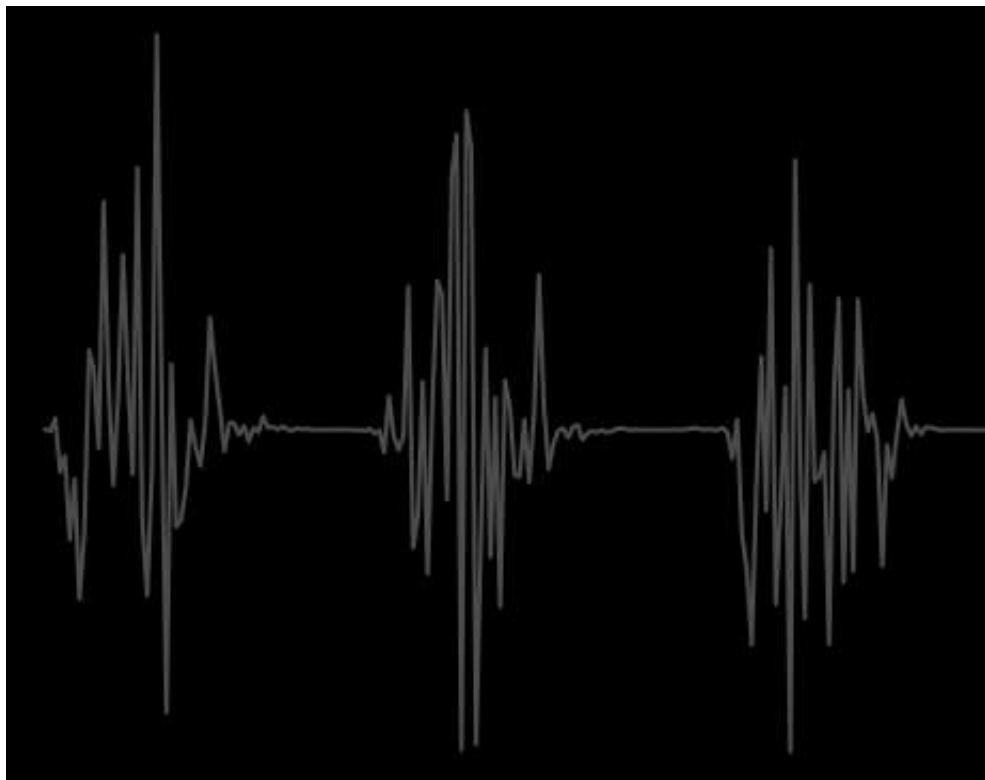
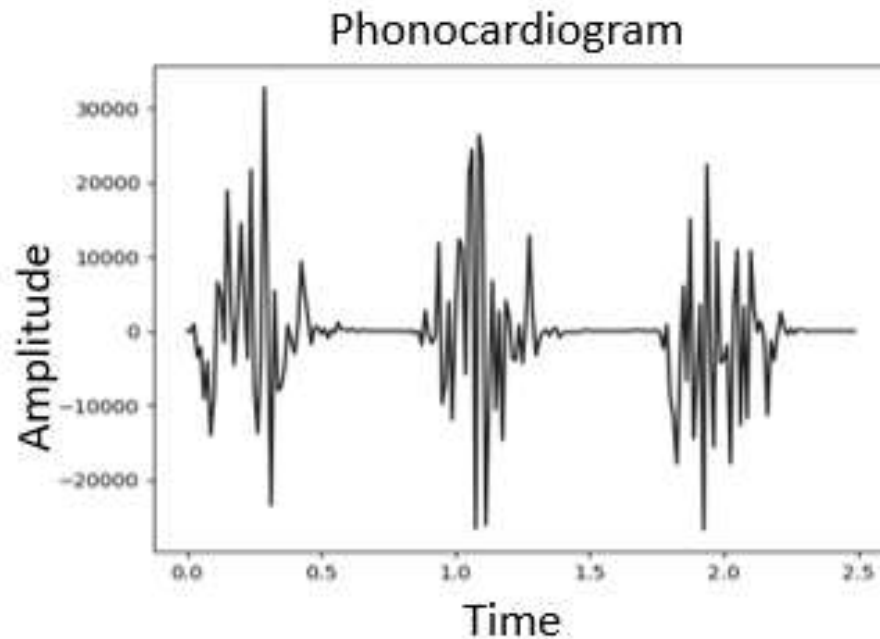


## Deep learning in Time Series Signal



# What is 1D Signal?

A 1D “Signal” describes how some physical quantity varies over time or space.



## Examples of Signals

- Sound pressure
- Radio broadcast
- Songs
- Electrocardiogram



# What is 1D Signal Processing?

**Manipulating a signal to change its characteristics or extract information**

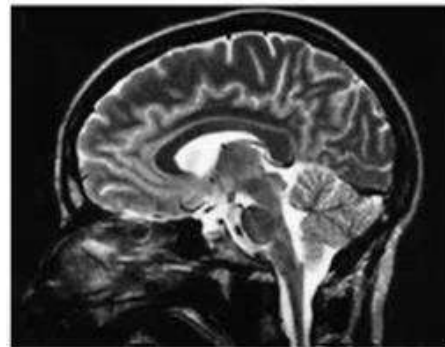
**Performed by:**

- Computer
- Special purpose integrated circuits
- Analog electrical circuits

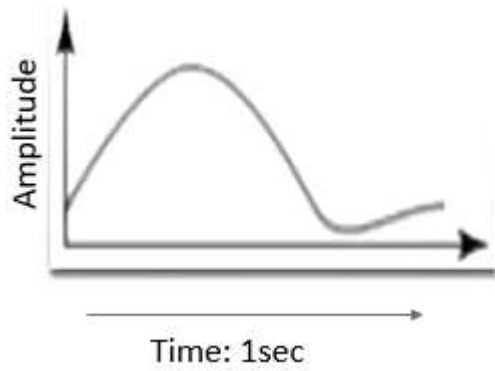


## Applications

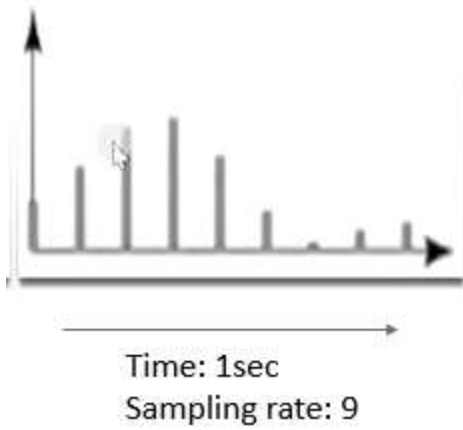
- Consumer electronics like HDTV, cell phones,...
- Transportation like GPS, engine control, tracking...
- Medical Monitoring like (EEG, ECG, EMG,...)
- Military like Target tracking, surveillance,...
- Remote sensing like Astronomy, climate monitoring, weather forecasting,...



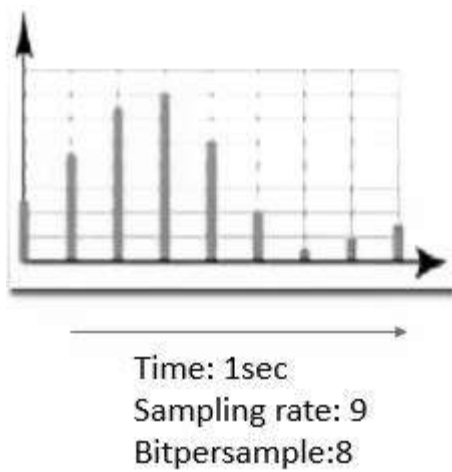
## PCM in Signal



Analog Signal

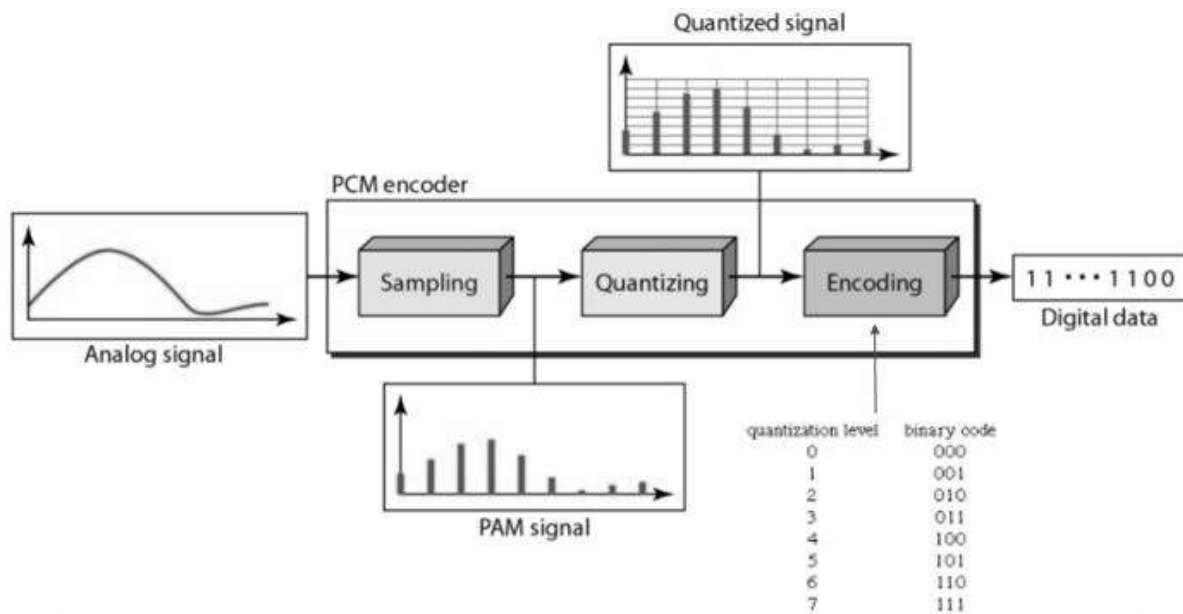


Sampled Signal



Quantized Signal

## PCM in Signal

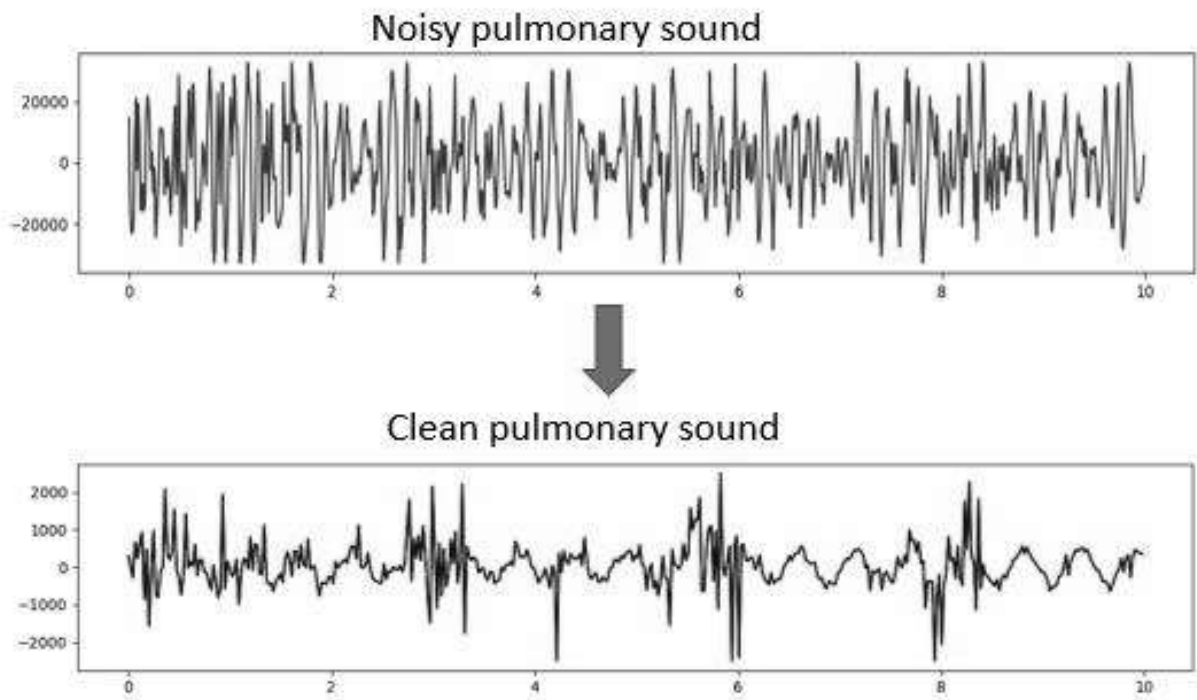


## Typical 1D Signal Processing Problems

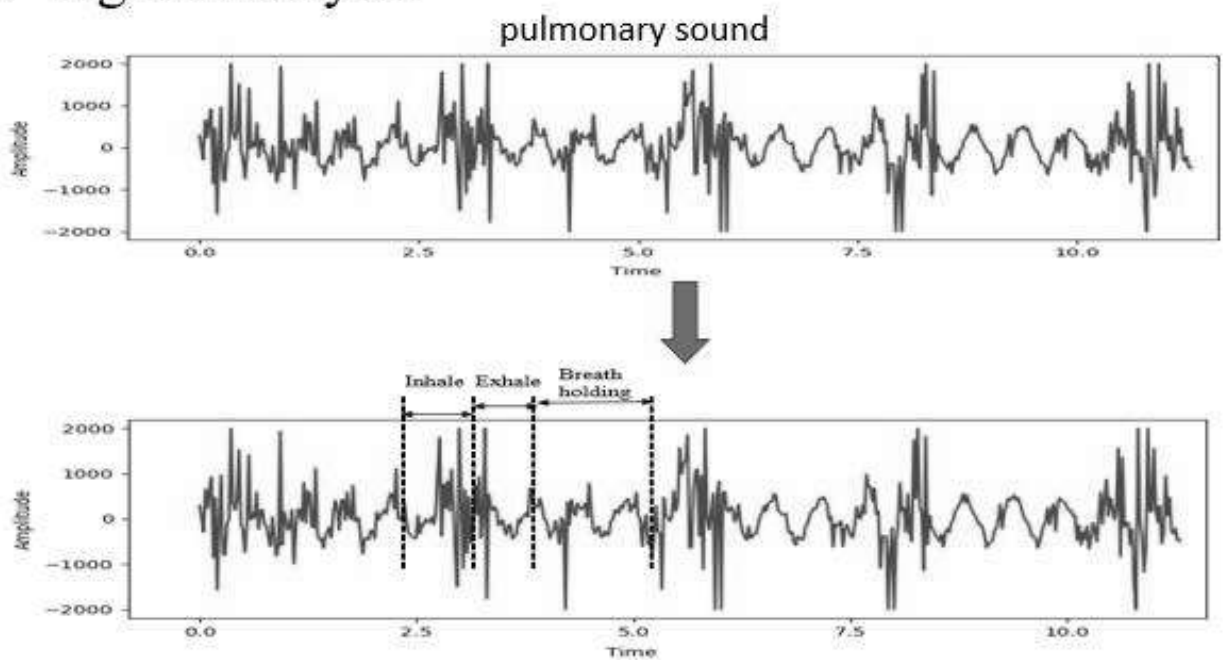
1. Signal Filtering
2. Signal Analysis
3. Extracting an indirect quantity from measured signal

## Typical 1D Signal Processing Problems

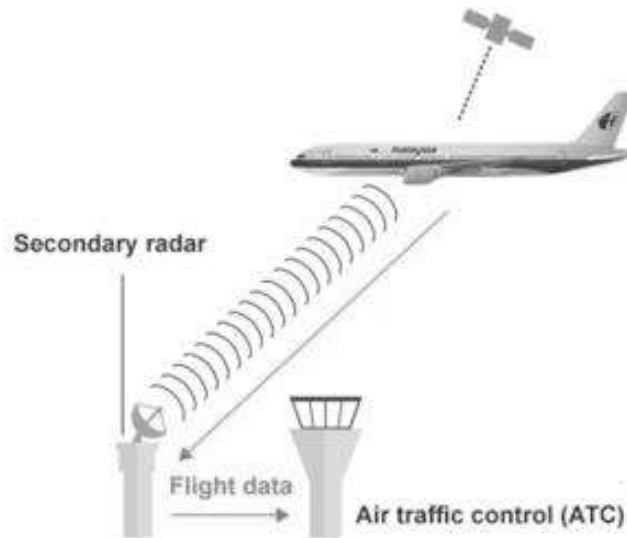
### 1. Signal Filtering



### 2. Signal Analysis



### 3. Extracting an indirect quantity from measured signal



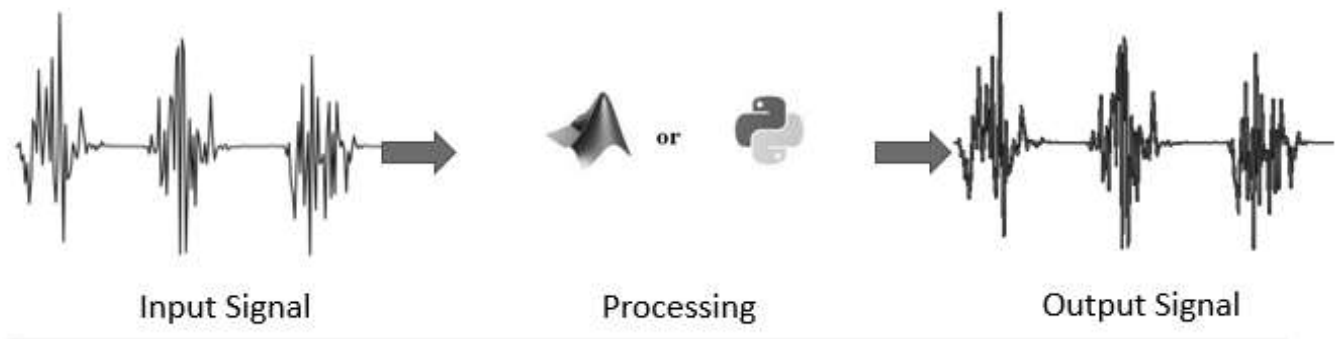
*E.g. Determining the aircraft position and velocity from radar signal*

### Signal Processing Philosophy

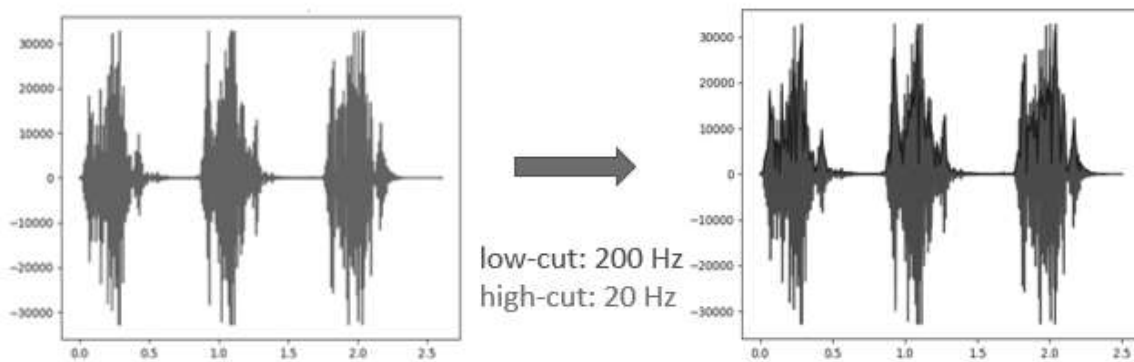
1. Model plays a fundamental role
  - Characterize “signal” and “noise”
  - Relate desired quantity to measured data
2. Model derives from prior knowledge
  - Physics, biology, etc.
3. Processing procedure based on models
4. Modeling issues
  - Poor model      Poor performance
  - Model complexity vs performance



## 1D Signal Processing

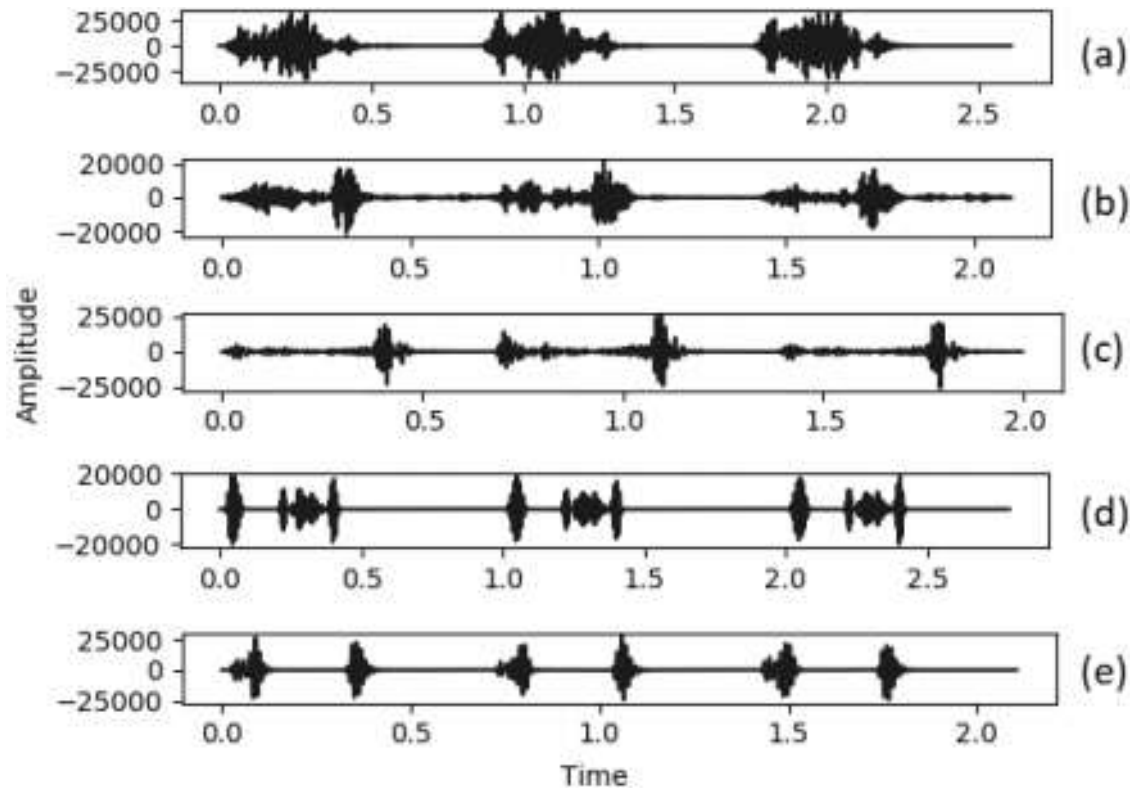


## 1D Signal Filtering



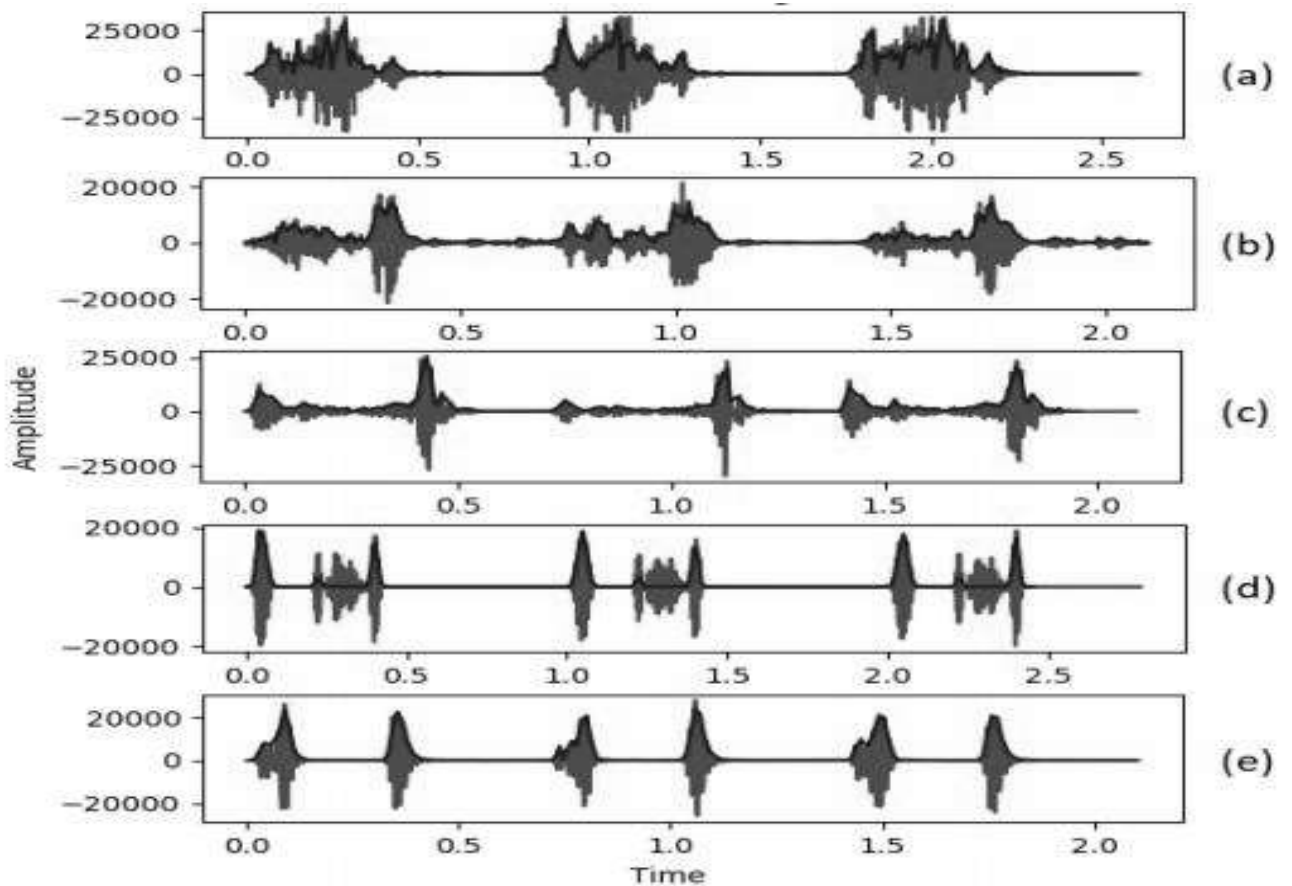
Represents the graph of Aortic Stenosis cardiac PCG signals amplitude with respect to time-domain before and after filtering. Also, it represents these signals with discriminating elements in a different colour. In these cardiographs, red colour denotes noise in these cardio signals, green colour denotes the extracted cardio signals after filtering and blue represents the peaks in the signals.

## Original PCG signals



**Fig.:** Represents the graph of PCG signals amplitude with respect to time-domain. (a) Graph of Aortic Stenosis cardiac PCG signal (b) Graph of Mitral Regurgitation cardiac PCG signal (c) Graph of Mitral Stenosis cardiac PCG signal (d) Graph of Mitral valve prolapse cardiac PCG signal (e) Graph of Normal cardiac PCG signal.

## Filtered PCG signals



**Fig:** Represents the graph of PCG signals amplitude with respect to time-domain after signal processing. Also, it represents these signals with discriminating elements in a different colour. In these cardiographs, red colour denotes noise in these cardio signals, green colour denotes the extracted cardio signals after signal-processing and blue represents the peaks in the signals. (a) Graph of Aortic Stenosis cardiac PCG signal (b) Graph of Mitral Regurgitation cardiac PCG signal (c) Graph of Mitral Stenosis cardiac PCG signal (d) Graph of Mitral valve prolapse cardiac PCG signal (e) Graph of Normal cardiac PCG signal.

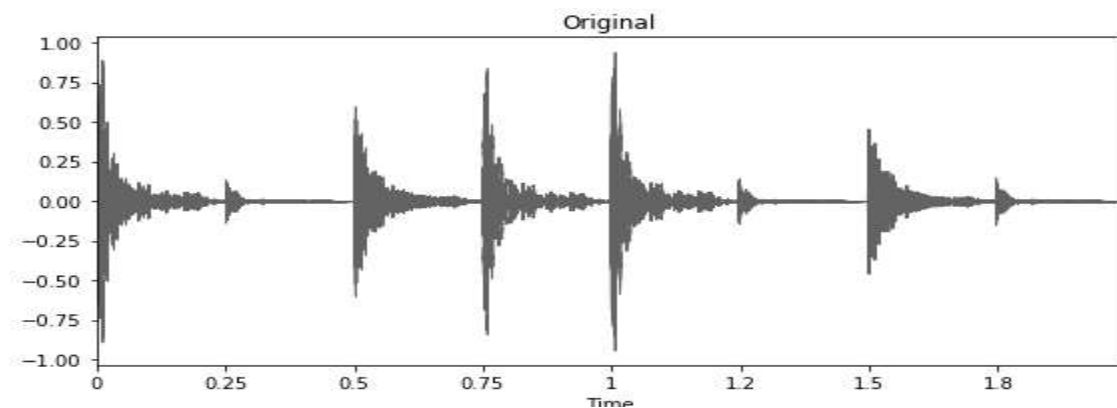
## 1D Signal Augmentation

**Data augmentation** is a strategy that enables practitioners to significantly increase the diversity of **data** available for training models, without actually collecting new **data**

### Major Types of 1D Data augmentation

1. Noise Injection
2. Shifting Time
3. Changing Pitch
4. Changing Speed

### Noise Injection



Represents the graph of 1D signals amplitude with respect to time-domain.

There is a given signal represented as  $I = [i_1, i_2, i_3, \dots, i_n]$

Same dimension of noise (random values) is selected

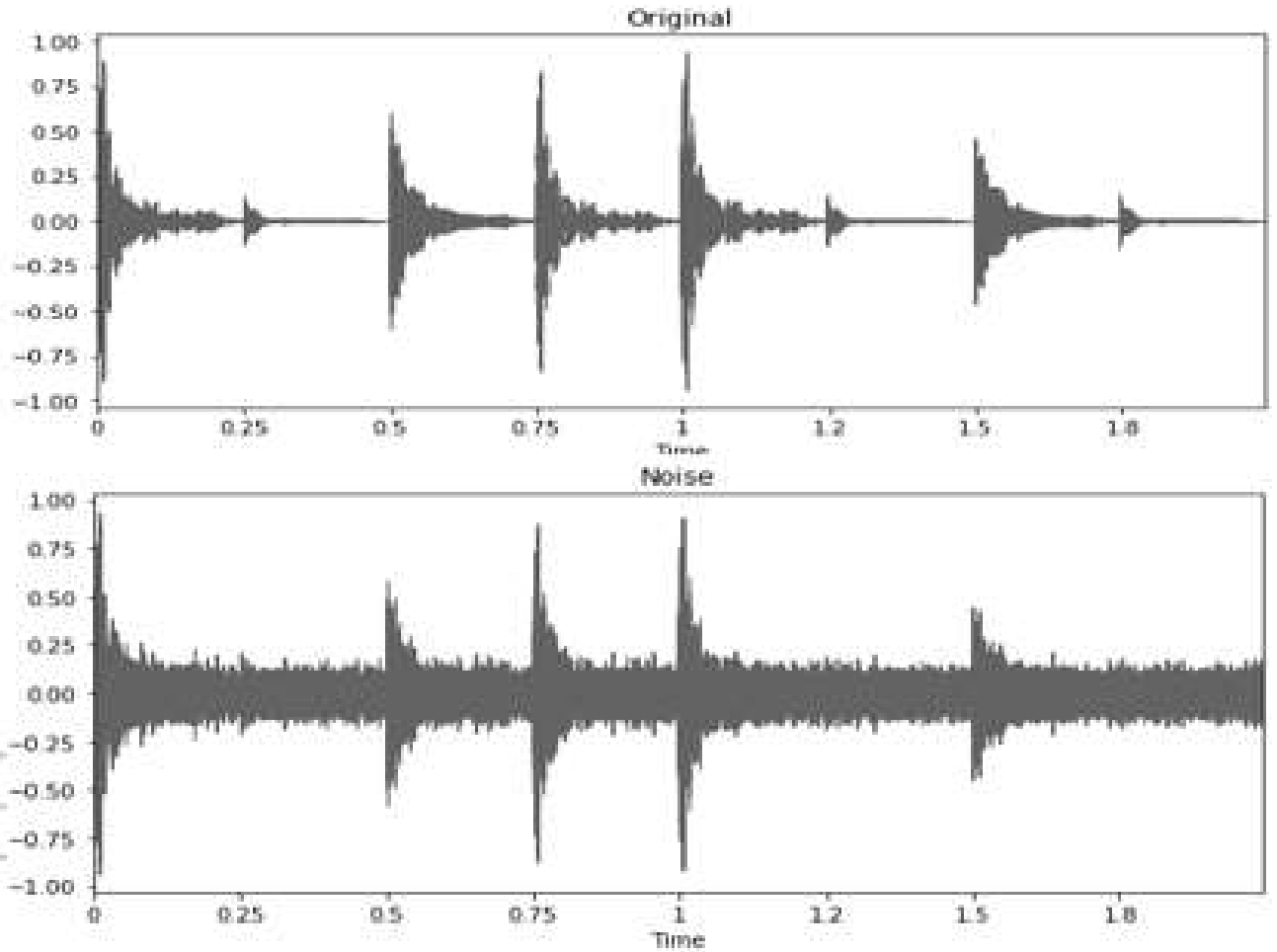
Represented as  $\theta = (\theta_1, \theta_2, \theta_3, \dots, \theta_n)$

Then, Augmentation signal  $\alpha$  is calculated

and their relation is  $\alpha = I + \theta * \lambda$ .

Where, deformations control parameter is represented as  $\lambda$

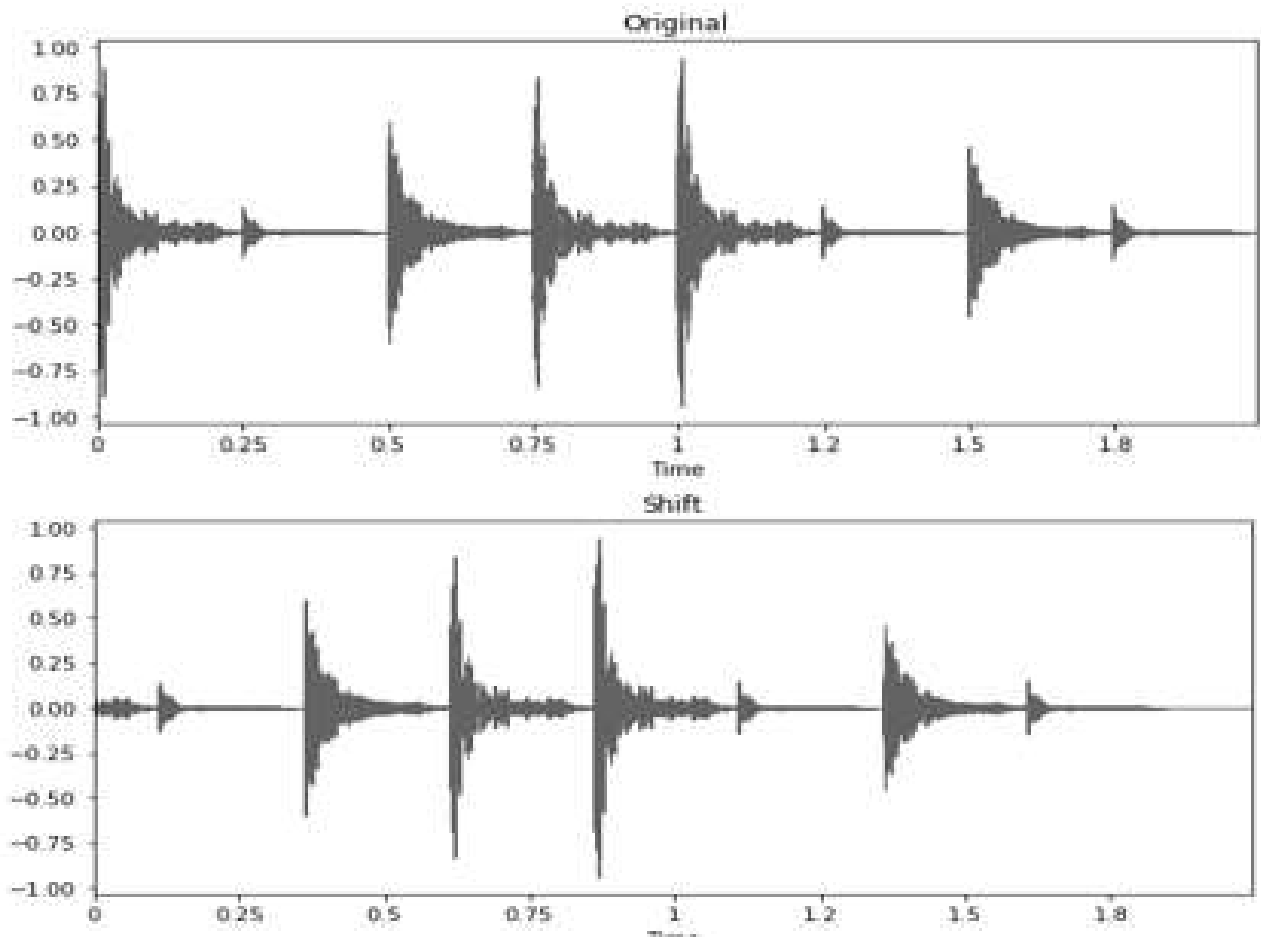
## 1. Noise Injection



Represents the graph of 1D signals amplitude with respect to time-domain before and after noise injection.

There is a given signal represented as  $I=[i_1, i_2, i_3, \dots, i_n]$  where  $i$  is the amplitude of the signal with the dimension of signal length  $n$ . Here the same dimension of background deformations is randomly selected and represented as  $\beta = (\beta_1, \beta_2, \beta_3, \dots, \beta_n)$  and deformations control parameter is represented as  $\lambda$ . From these audio deformations,  $\alpha$  is calculated and their relation is  $\alpha = I + \beta * \lambda$ . Where, background deformations  $\beta$  belong to the interval  $(0, 1)$  and deformations control parameter  $\lambda$  is taken as 1000 Hz to convert the deformation value into three-digit value.

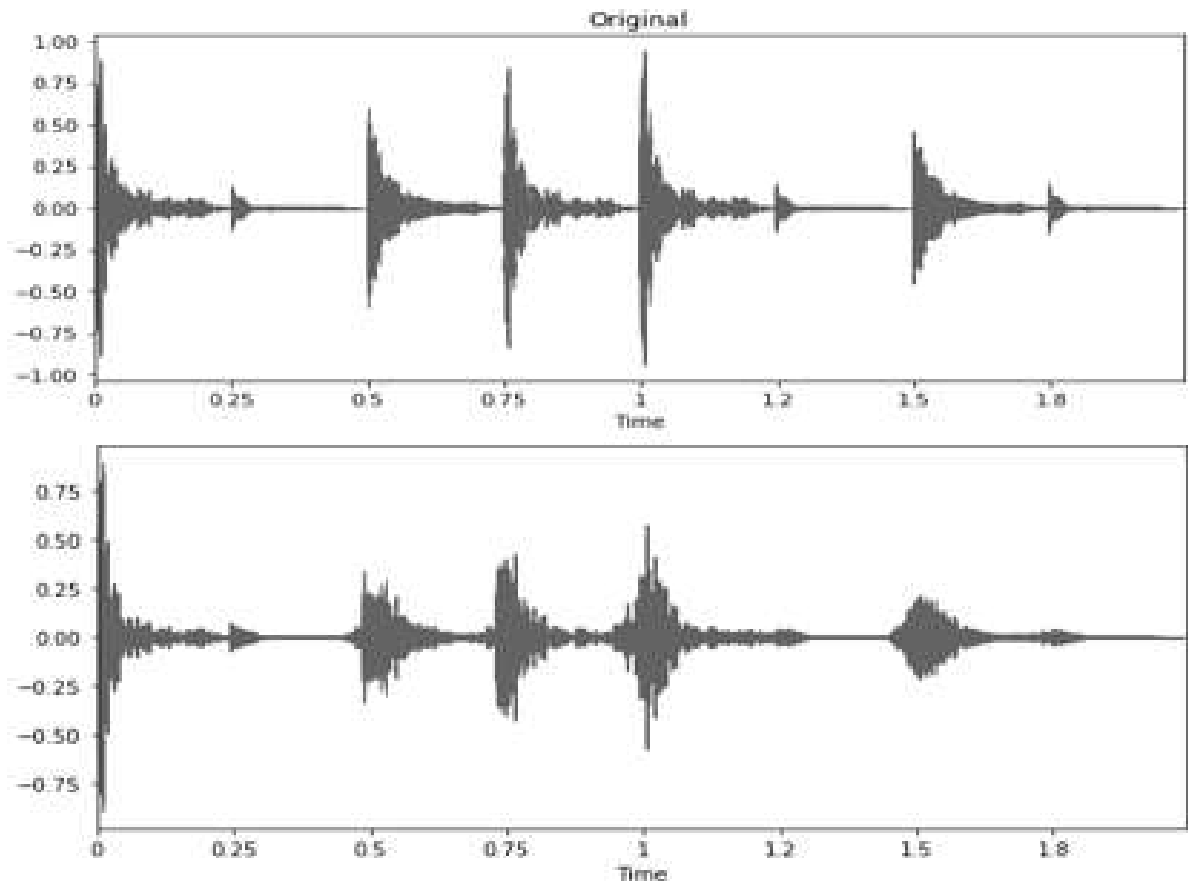
## 2. Shifting Time



Represents the graph of 1D signals amplitude with respect to time-domain before and after shifting time

There is a given signal represented as  $I = [i_1, i_2, i_3, \dots, i_n]$  where  $i$  is the amplitude of the signal with the dimension of signal length  $n$ . Here the randomly selected and represented as  $\beta = (\beta_1, \beta_2, \beta_3, \dots, \beta_n)$  of maximum time shift \* sampling rate and From these time shifted signal,  $\alpha$  is calculated and their relation is  $\alpha = I + \beta$  for left shift.

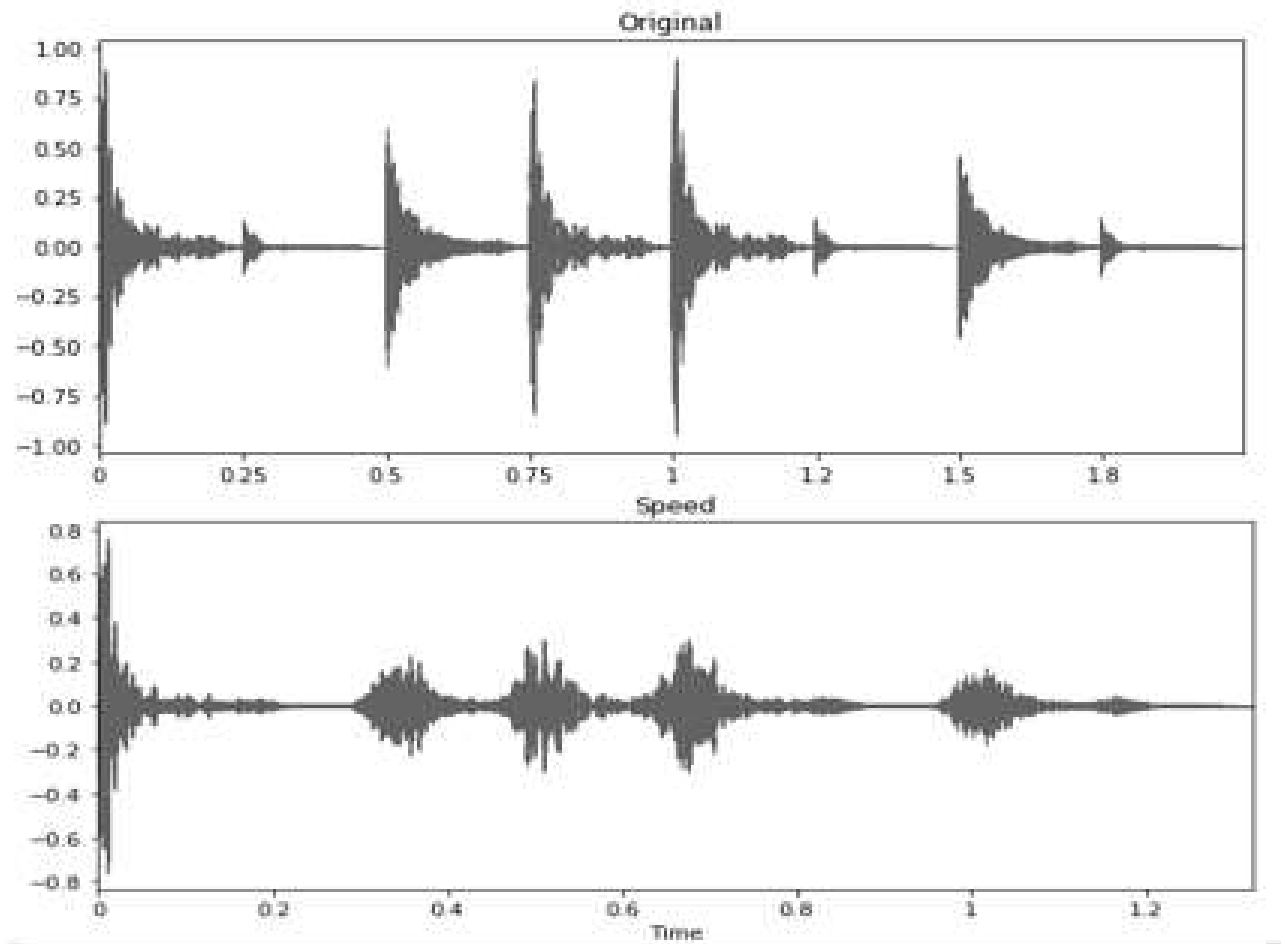
### 3. Changing Pitch



Represents the graph of 1D signals amplitude with respect to time-domain before and after changing pitch

There is a given signal represented as  $I=[i_1, i_2, i_3, \dots, i_n]$  where  $i$  is the amplitude of the signal with the dimension of signal length  $n$ . Here the pitch control parameter is represented as  $\lambda$  and sampling rate as  $s$ . From these time shifted signal,  $\alpha$  is calculated and their relation is  $\alpha = fx(I, s, \lambda)$ .

## 4. Changing Speed



Represents the graph of 1D signals amplitude with respect to time-domain before and after changing speed.

There is a given signal represented as  $I=[i_1, i_2, i_3, \dots, i_n]$  where  $i$  is the amplitude of the signal with the dimension of signal length  $n$ . Here the speed control parameter is represented as  $\lambda$ . From these time shifted signal,  $\alpha$  is calculated and their relation is  $\alpha = fx(I, \lambda)$ .



## **1D Signal Features extraction**

Feature extraction is used to find important points for discriminating the data among the different classes from the input data.

Feature extraction increases the accuracy of learned models by extracting features from the input data.

### **Statistical-based feature extraction**

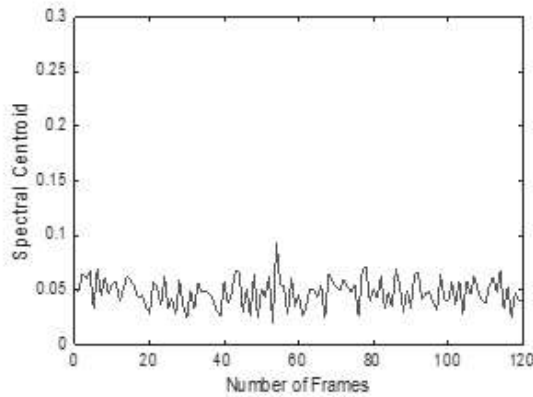
1. Spectral Centroid
2. Energy Entropy
3. Spectral Roll off
4. Spectral Flux
5. Volume
6. Zero Crossing Rate

## 1. Spectral Centroid

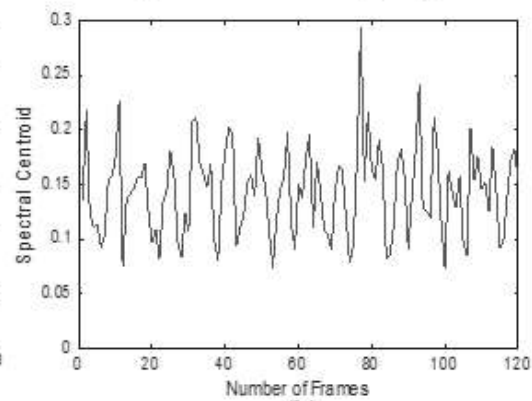
$$\text{Centroid} = \frac{\sum_{n=0}^{N-1} f(n) x(n)}{\sum_{n=0}^{N-1} x(n)}$$

$x(n)$  is the weighted frequency of bin number

$f(n)$  is the center frequency of bin



(a)

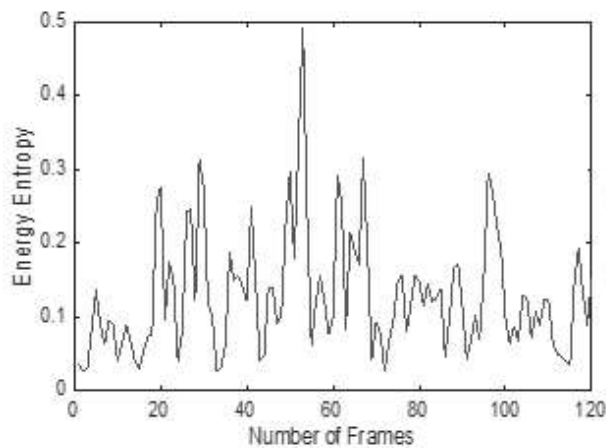


(b)

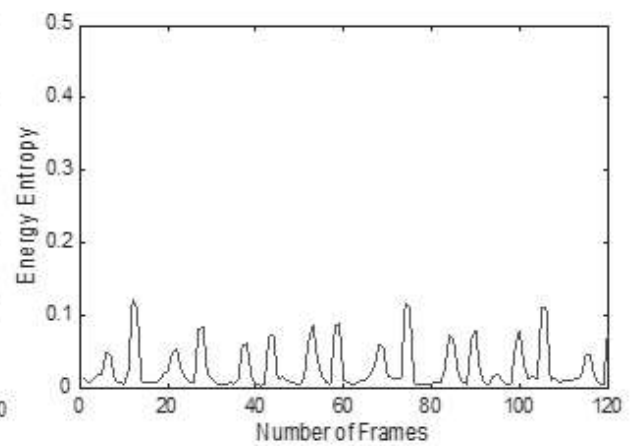
Spectral Centroid: (a) Normal heart sound (b) Abnormal heart Sound

## 2. Energy Entropy

$$I_j = -\sum_{i=1 \dots k} \sigma_i^2 \log_2 \sigma_i^2$$



(a)

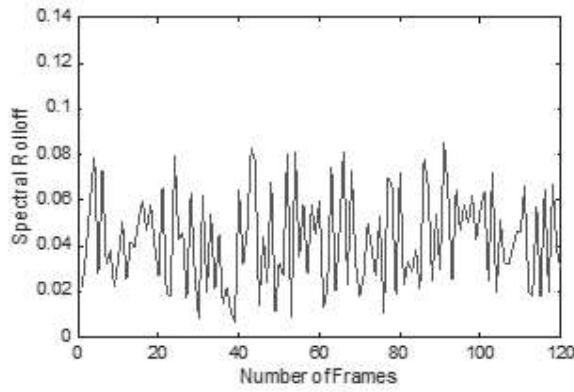


(b)

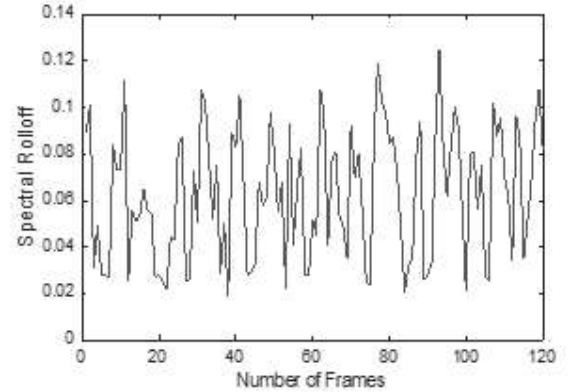
Energy Entropy : (a) Normal heart sound (b) Abnormal heart Sound

### 3. Spectral Roll off

$$\sum_{n=1}^{R_t} M_t[n] = 0.85 \sum_{n=1}^N M_t[n]$$



(a)

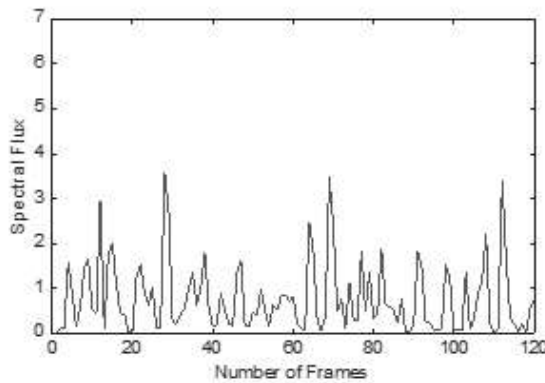


(b)

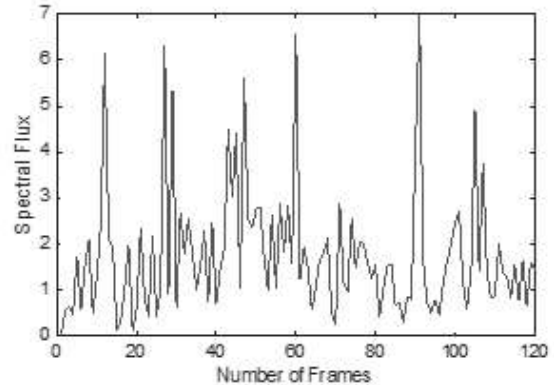
Spectral Roll off : (a) Normal heart sound (b) Abnormal heart Sound

### 4. Spectral Flux

$$F_j = \sum_{k=0 \dots S-1} (N_{j,k} - N_{j-1,k})^2$$



(a)

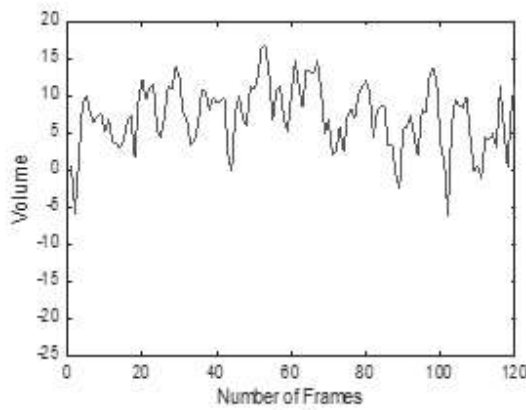


(b)

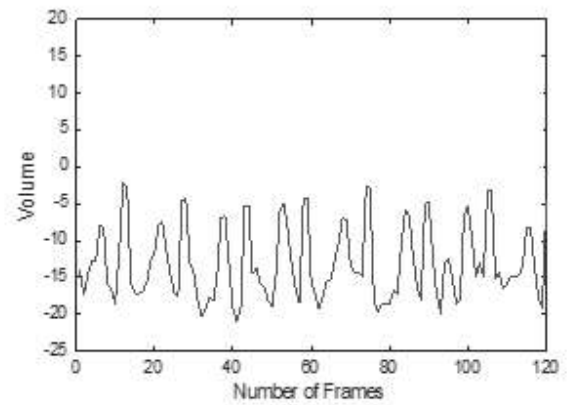
Spectral Flux : (a) Normal heart sound (b) Abnormal heart Sound

## 5. Volume

$$volume = 10 * \log_{10} \sum_{i=1}^n s_i^2$$



(a)



(b)

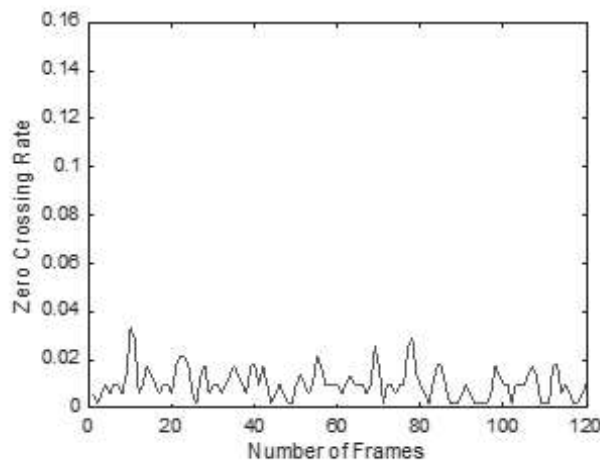
---

Spectral Flux : (a) Normal heart sound (b) Abnormal heart Sound

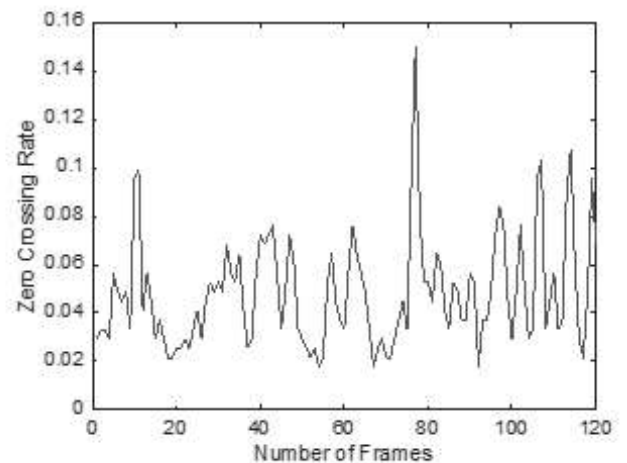
---

## 6. Zero Crossing Rate

$$Z_j = \frac{1}{2S} \sum_{i=1 \dots S} |sgn(x_i) - sgn(x_{i-1})|$$



(a)



(b)

---

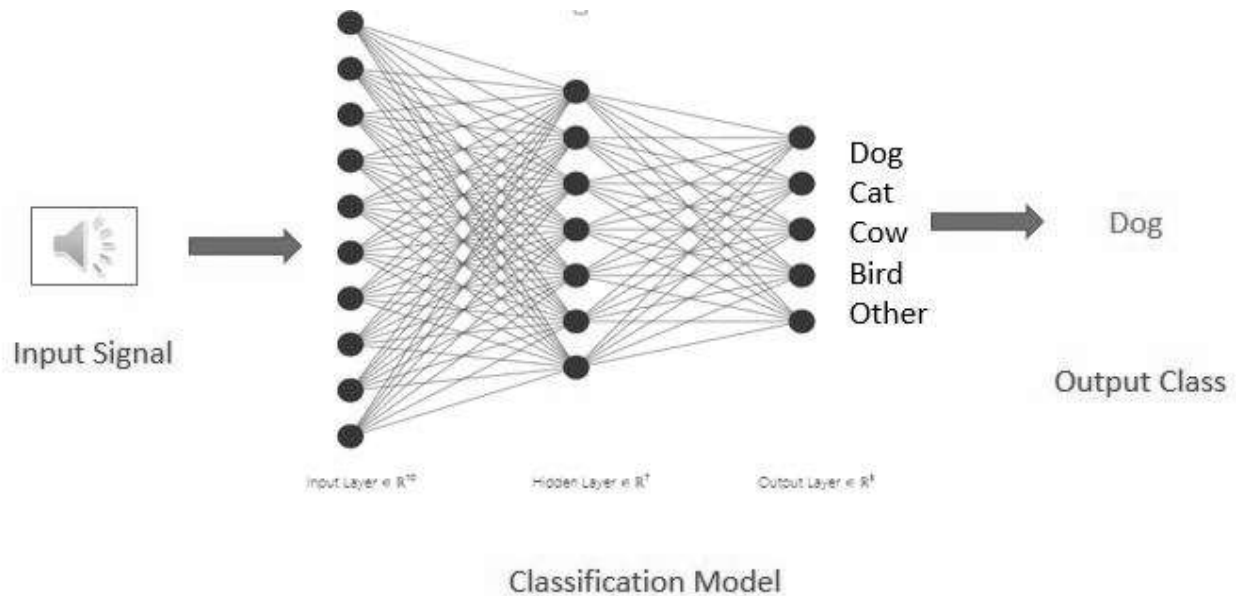
Zero Crossing Rate : (a) Normal heart sound (b) Abnormal heart Sound

---

Other types of features used in 1D signal

1. Fourier transform
2. Spectrogram
3. Sort-Time Fourier Transform
4. Mel Filter Bank
5. Mel Frequency Cepstral Coefficients

## Deep learning in 1D



## Convolution in Time Series Signal

Inverted Kernel

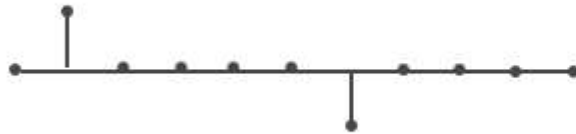
1 2 .5

Convolution Operation

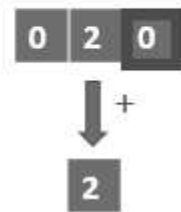
$$.5 \times 0 = 0$$

Time series Signal

0 1 0 0 0 0 -1 0 0 0 0



Time series Signal Representation



## Convolution in Time Series Signal

Time series Signal

0 1 0 0 0 0 -1 0 0 0 0



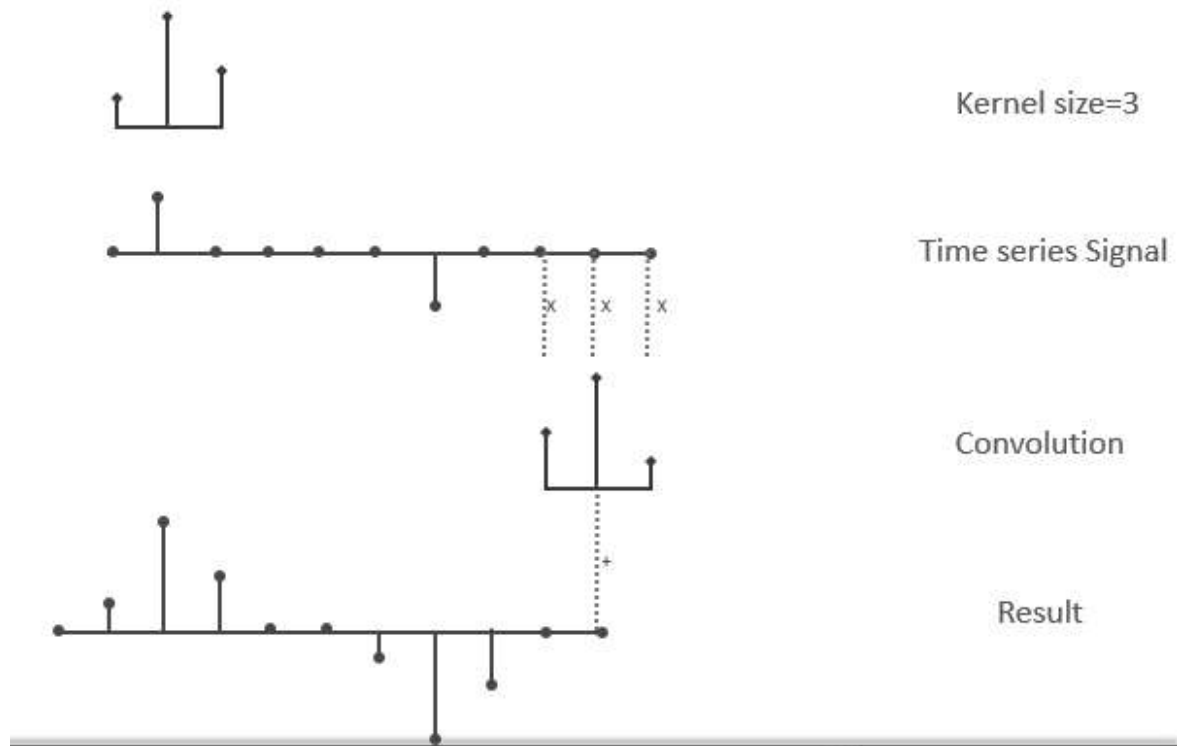
Inverted Kernel

1 2 .5

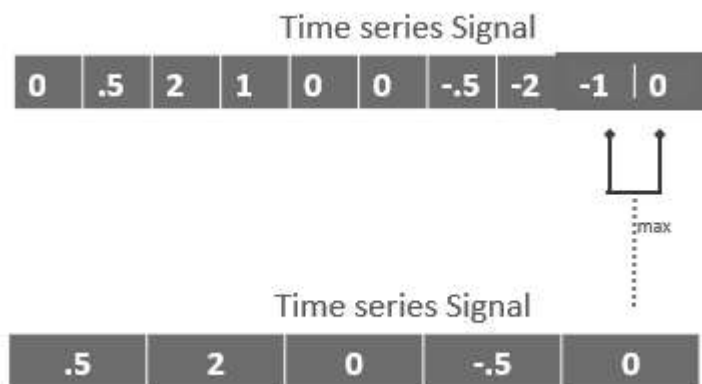
=

Time series Signal

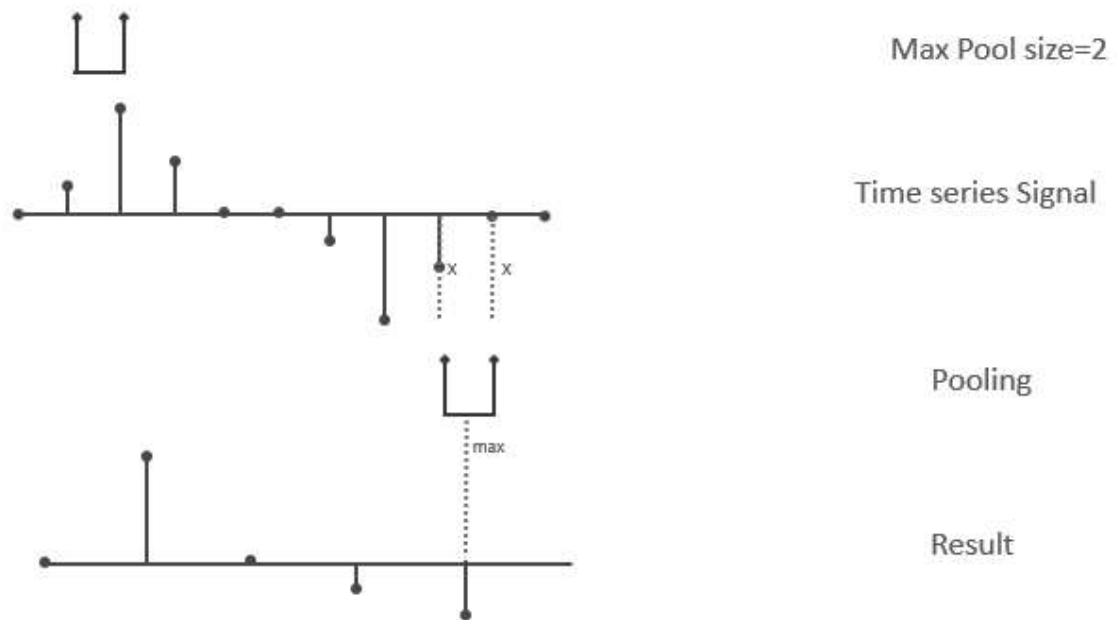
0 .5 2 1 0 0 -0.5 -2 -1 0 0



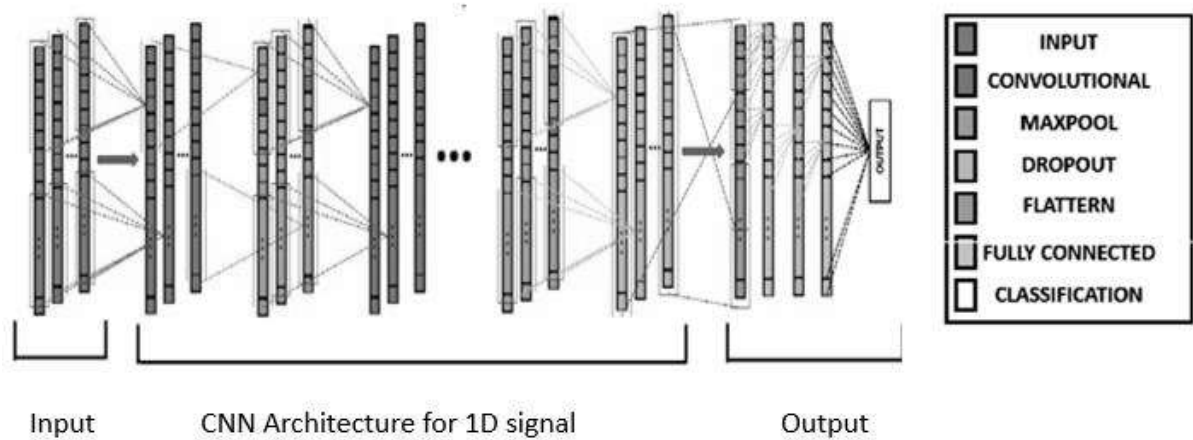
## Max Pooling in Time Series Signal



## Max Pooling in Time Series Signal



## Time Series Signal Classifications using CNN





# Deep learning for Signal Classifications using CNN

There is a given signal represented as  $I=[i_1, i_2, i_3, \dots, i_n]$  where  $i$  is the amplitude of the signal with the dimension of signal length 20000. It is classified into 5 classes.

```
def cnn_model(size):
    m = Sequential()
    m.add(Conv1D(filters=8, kernel_size=3, strides=3, padding='valid', activation='relu', input_shape=size))
    m.add(Conv1D(filters=16, kernel_size=3, strides=3, padding='valid', activation='relu'))
    m.add(Conv1D(filters=32, kernel_size=3, strides=3, padding='valid', activation='relu'))
    m.add(Conv1D(filters=64, kernel_size=3, strides=3, padding='valid', activation='relu'))
    m.add(Conv1D(filters=128, kernel_size=3, strides=3, padding='valid', activation='relu'))
    m.add(MaxPooling1D(pool_size=2))
    m.add(Dropout(0.5))
    m.add(Flatten())
    m.add(Dense(128, activation='relu'))
    m.add(Dense(5, activation='softmax'))
    m.compile(loss='categorical_crossentropy', optimizer='adam', metrics=['accuracy'])
    m.summary()
    return m

model=cnn_model((20000,1))
```

Model: "sequential\_8"

Layer (type)	Output Shape	Param #
conv1d_38 (Conv1D)	(None, 6666, 8)	32
conv1d_39 (Conv1D)	(None, 2222, 16)	400
conv1d_40 (Conv1D)	(None, 740, 32)	1568
conv1d_41 (Conv1D)	(None, 246, 64)	6208
conv1d_42 (Conv1D)	(None, 82, 128)	24704
max_pooling1d_8 (MaxPooling1D)	(None, 41, 128)	0
dropout_9 (Dropout)	(None, 41, 128)	0
flatten_8 (Flatten)	(None, 5248)	0
dense_15 (Dense)	(None, 128)	671872
dense_16 (Dense)	(None, 5)	645
Total params: 705,429		
Trainable params: 705,429		
Non-trainable params: 0		

# Deep learning for Signal Classifications using RNN

There is a given signal represented as  $I=[i_1, i_2, i_3, \dots, i_n]$  where  $i$  is the amplitude of the signal with the dimension of signal length 20000. It is classified into 5 classes.

```
def rnn_model(size):
    m = Sequential()
    m.add(LSTM(128, return_sequences=True, input_shape=size))
    m.add(LSTM(128, return_sequences=True))
    m.add(Dropout(0.5))
    m.add(TimeDistributed(Dense(128, activation='relu'))))
    m.add(TimeDistributed(Dense(16, activation='relu'))))
    m.add(Flatten())
    m.add(Dense(128, activation='relu'))
    m.add(Dense(5, activation='softmax'))
    m.compile(loss='categorical_crossentropy', optimizer='adadelta', metrics=['accuracy'])
    m.summary()
    return m

model=rnn_model((20000,1))
```

Model: "sequential\_14"

Layer (type)	Output Shape	Param #
lstm_9 (LSTM)	(None, 20000, 128)	66560
lstm_10 (LSTM)	(None, 20000, 128)	131584
dropout_14 (Dropout)	(None, 20000, 128)	0
time_distributed_8 (TimeDistributed)	(None, 20000, 128)	16512
time_distributed_9 (TimeDistributed)	(None, 20000, 16)	2064
flatten_12 (Flatten)	(None, 320000)	0
dense_30 (Dense)	(None, 128)	40960128
dense_31 (Dense)	(None, 5)	645
Total params: 41,177,493		
Trainable params: 41,177,493		
Non-trainable params: 0		