

## Design of a coherent sensor signal sampling methodology to implement a data logger to send audio sensor data from STM32 to NanoEdge AI Studio on PC

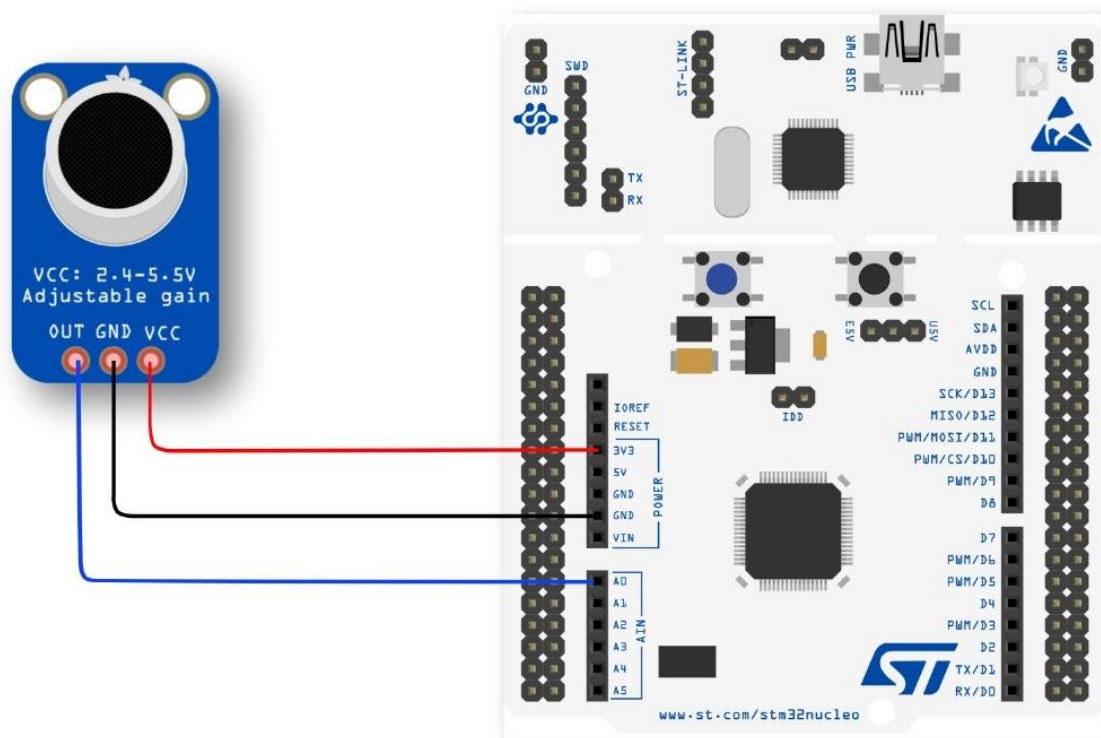
### Objective:

The objective of this experiment is to interface an audio sensor to an STM32 microcontroller and create a datalogger code. This datalogger code will create a buffer where all the audio data sample will be stored, using which we will be able to create datasets of audio samples to build a machine learning model in the NanoEdge AI Studio.

### Requirements:

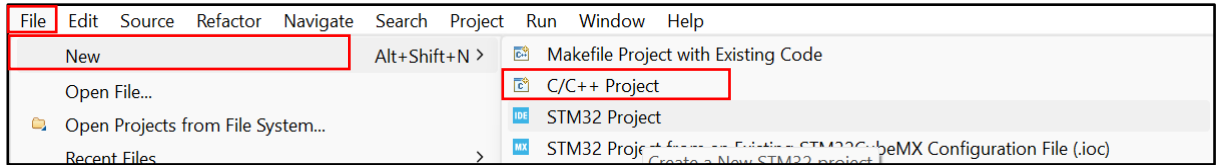
1. STM32 Cube IDE software.
2. Audio Sensor (Analog).
3. STM32 Microcontroller.
4. USB Cable for the microcontroller.
5. Jumper Wires.
6. PC or Laptop.

### Connection Diagram:

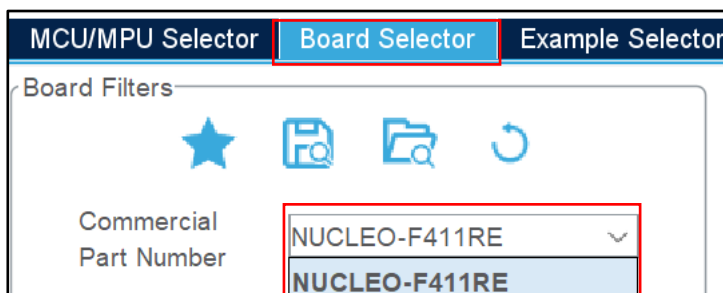


## Procedure:

1. Click on **File**→**New**→**STM32 Project** to start your project on Cube IDE.



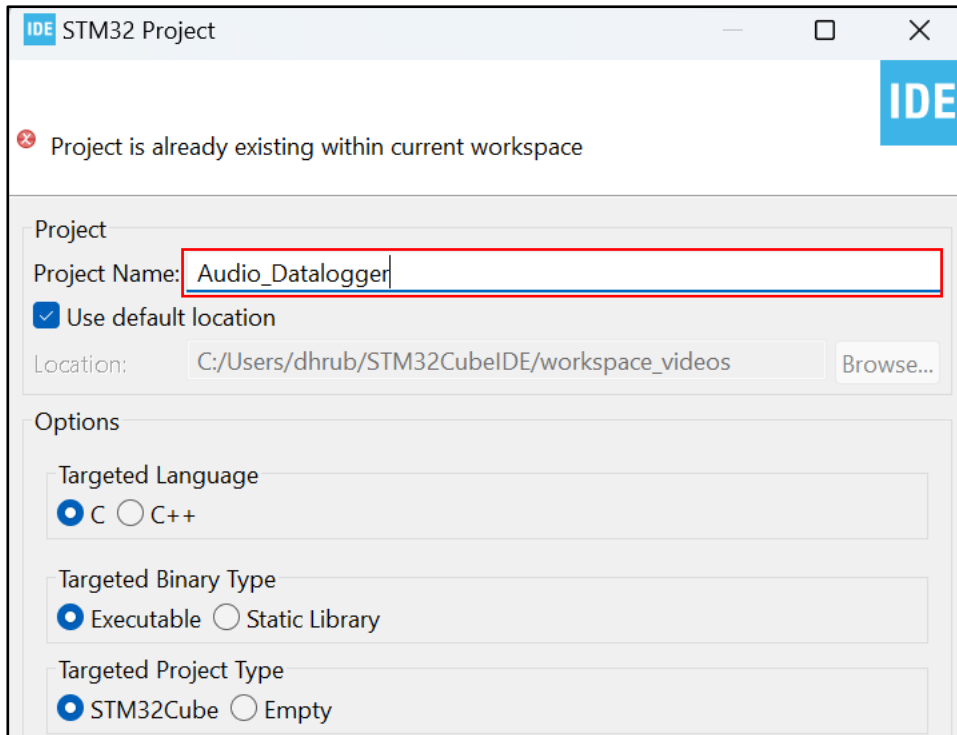
2. A **Target Selection** window will open. Click on **Board Selector**, where you need to select the microcontroller board you are working with.



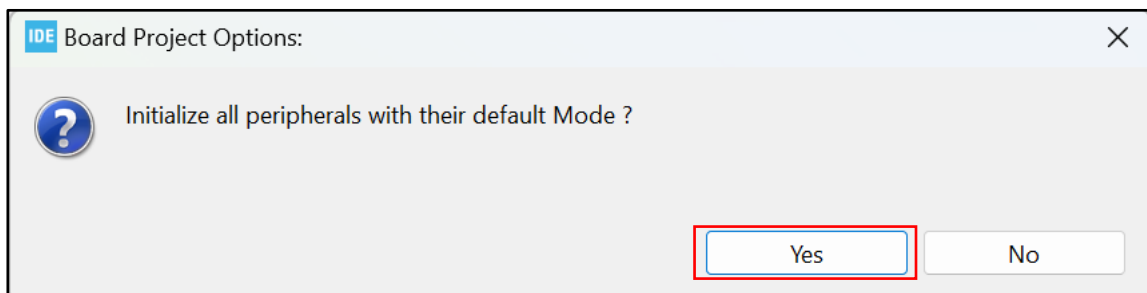
3. After this on the right-hand side of the window, under **Board List** you will see the board you have selected. Click on the board and then click on **Next**.



4. In the next window you need to give your project a name, rest of the things will remain by default as it is for now. Click on **Next**.



5. Cube IDE will ask if you want to initialize all peripherals with their default mode, click on **Yes**.



6. Next on the left-hand side **Categories** → **Analog** select **ADC1** then select **IN0** under **Mode**. Under **Configuration** select **Parameter Settings**, change **Resolution** to 10-bits, then change the **External Trigger ConversionSource** to **Timer 2 Trigger Out Event** and **External Trigger Conversion Mode** to **Trigger detection on the falling edge**. Under **NVIC** enable the **ADC1 global interrupt**. Otherwise, you can also go to **System Core** → **NVIC** and enable the same.

Analog

ADC1

ADC1 Mode and Configuration

Mode	
<input checked="" type="checkbox"/>	IN0
<input type="checkbox"/>	IN1
<input type="checkbox"/>	IN2
<input type="checkbox"/>	IN3
<input type="checkbox"/>	IN4
<input type="checkbox"/>	IN5
<input type="checkbox"/>	IN7

ADC\_Settings

Clock Prescaler	PCLK2 divided by 2
Resolution	10 bits (13 ADC Clock cycles)
Data Alignment	12 bits (15 ADC Clock cycles)
Scan Conversion Mode	10 bits (13 ADC Clock cycles)
Continuous Conversion Mo...	8 bits (11 ADC Clock cycles)
Discontinuous Conversion ...	6 bits (9 ADC Clock cycles)

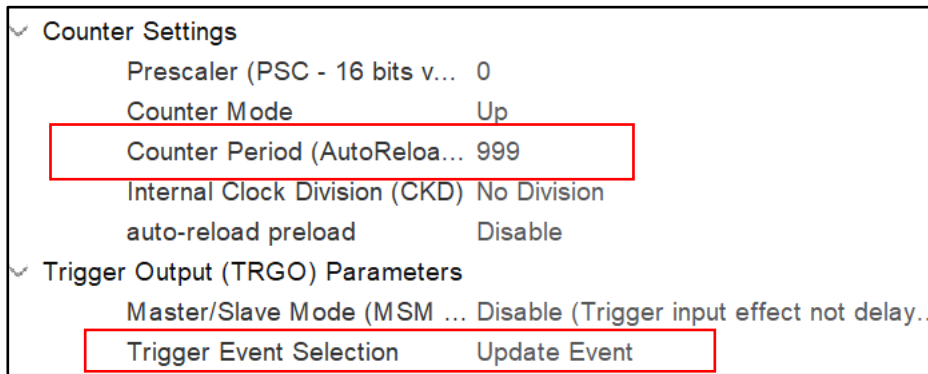
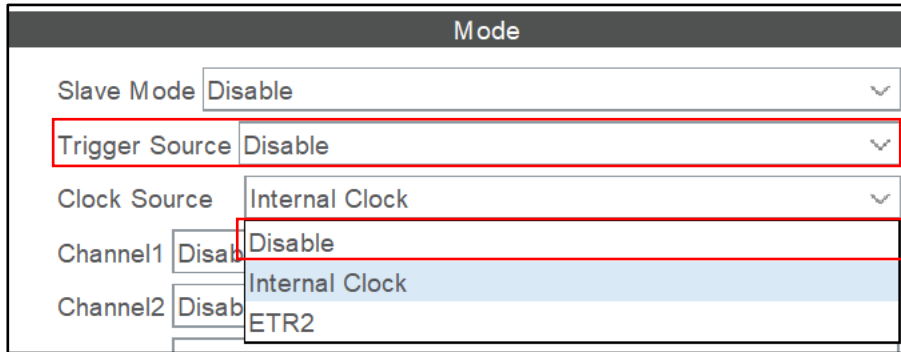
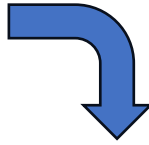
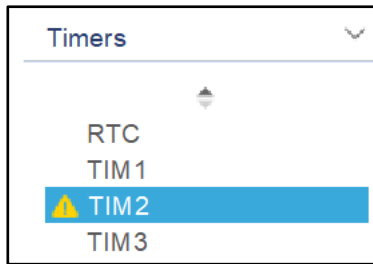
Search (Ctrl+F)

Continuous Conversion Mo...	Timer 2 Capture Compare 4 event
Discontinuous Conversion ...	Timer 2 Trigger Out event
DMA Continuous Requests	Timer 3 Capture Compare 1 event
End Of Conversion Selection	Timer 3 Trigger Out event
ADC_Regular_ConversionMode	Timer 4 Capture Compare 4 event
Number Of Conversion	Timer 5 Capture Compare 1 event
External Trigger Conversio...	Timer 5 Capture Compare 2 event
	Timer 5 Capture Compare 3 event
	Timer 2 Trigger Out event

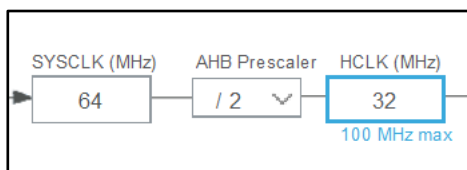
External Trigger Conversio...	Trigger detection on the rising edge
Rank	Trigger detection on the rising edge
ADC_Injected_ConversionMode	Trigger detection on the falling edge
Number Of Conversions	Trigger detection on both

NVIC Interrupt Table	Enabled	Preemption Priority	Sub Priority
ADC1 global interrupt	<input checked="" type="checkbox"/>	0	0

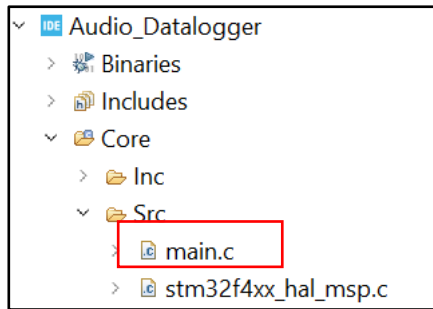
7. Next go to **Timer**, select **TIM2** and select **Clock Source** as **Internal Clock**. Update the **Counter Period** value as **999** and change **Trigger Event Selection** to **Update Event**.



8. Go to **Clock Configuration**, enter 32 as a value in **HCLK** and press enter to configure the peripheral and timer clocks. In this experiment we are trying to sample the audio signals at 32 KHz.



9. Press **Ctrl+S** to generate your code. On the left-hand side of the Cube IDE, under **Project Explorer** go to the project you have created (For example I have named my project as (Audio\_Datalogger)**Audio\_Datalogger**→**Core**→**Src**→**main.c** (double click to load the code).



10. Cube IDE automatically generates a code format based on the configurations you have done. Cube IDE uses HAL libraries. Below is the code snippets, please put your code in the appropriate places in the **main.c** file.

```
22/* Private includes -----*/
23/* USER CODE BEGIN Includes */
24#include "stdio.h"
25#include "string.h"
26/* USER CODE END Includes */
```

```
34/* USER CODE BEGIN PD */
35#define DATA_INPUT_USER 256
36#define AXIS_NUMBER 1
37/* USER CODE END PD */
```

```
51/* USER CODE BEGIN PV */
52float mic_x = 0.0;
53float mic_buffer[DATA_INPUT_USER * AXIS_NUMBER] = {0};
54volatile uint32_t buffer_index = 0;
55/* USER CODE END PV */
```

```
63/* USER CODE BEGIN PFP */
64void Log(void);
65void fill_mic_buffer(void);
66int __io_putchar(int);
67/* USER CODE END PFP */
```

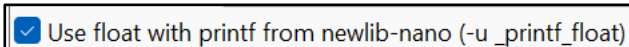
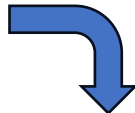
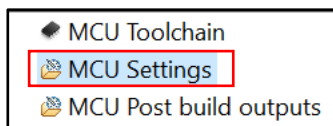
```
70/* Private user code -----*/
71/* USER CODE BEGIN 0 */
72void HAL_ADC_ConvCpltCallback(ADC_HandleTypeDef* hadc)
73{
74    if (hadc->Instance == ADC1)
75    {
76        mic = HAL_ADC_GetValue(&hadc1);
77        mic_buffer[buffer_index] = mic;
78        buffer_index++;
79        if (buffer_index >= DATA_INPUT_USER)
80        {
81            buffer_index = 0; // Reset index to 0 when it reaches the buffer size
82        }
83    }
84}
85}
86/* USER CODE END 0 */
```


```
119  /* USER CODE BEGIN 2 */
120 HAL_ADC_Start_IT(&hadcl1);
121 HAL_TIM_Base_Start(&htim2);
122  /* USER CODE END 2 */
```

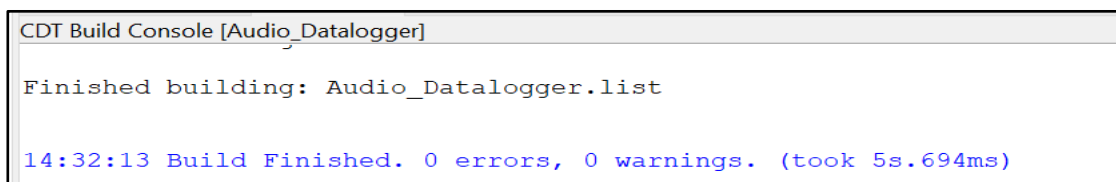
```
139  /* USER CODE BEGIN WHILE */
140  while (1)
141  {
142      Log();
143  } /* USER CODE END WHILE */
```


```
363 /* USER CODE BEGIN 4 */
364 int __io_putchar(int ch) {
365     HAL_UART_Transmit(&huart2, (uint8_t *)&ch, 1, 0xFFFF);
366     return ch;
367 }
368 void fill_mic_buffer() {
369     for(int i=0; i<DATA_INPUT_USER; i++) {
370         mic_buffer[AXIS_NUMBER * i] = mic_x;
371         HAL_Delay(3);
372     }
373 }
374 void Log() {
375     fill_mic_buffer();
376     for(int i=0; i<DATA_INPUT_USER; i++) {
377         printf("%.2f", mic_buffer[AXIS_NUMBER * i]);
378         printf(" ");
379     }
380     printf("\r\n");
381 }
```

11. Right click on the Audio\_Classification project and select **Properties**. Go to **C/C++ Build** → **Settings**. Next select **MCU Settings** and enable the option **Use float with printf from newlib-nano (-u\_printf\_float)**.

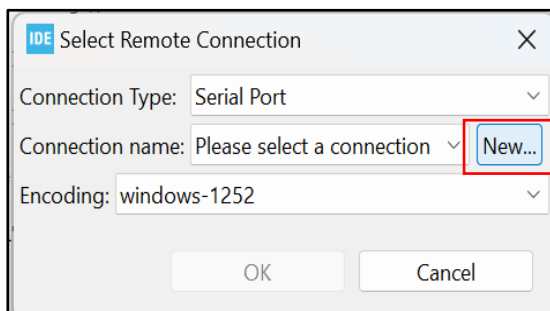
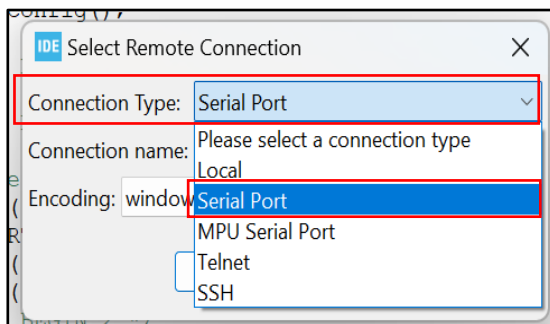
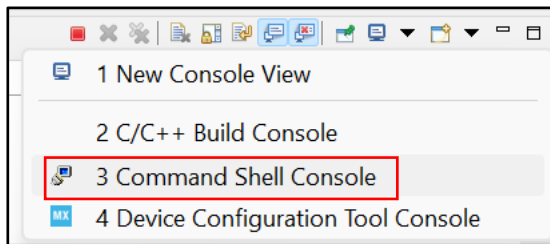
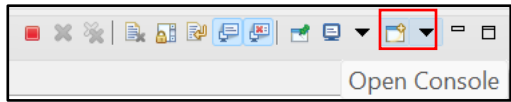


12. Now click on the build  symbol on the top left corner on your Cube IDE. If you have done everything correctly your code should be built without any errors.

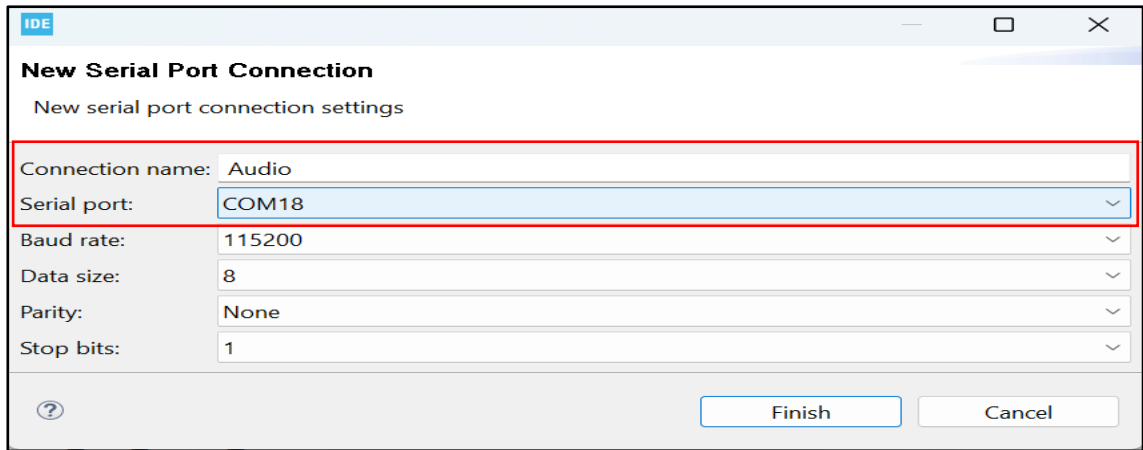



13. Next connect your STM32 board with your audio sensor connect to it to your PC and click on the **Debug**  icon to start the Debugging process. An **Edit Configuration** window will open, click on **OK**, without making any changes.

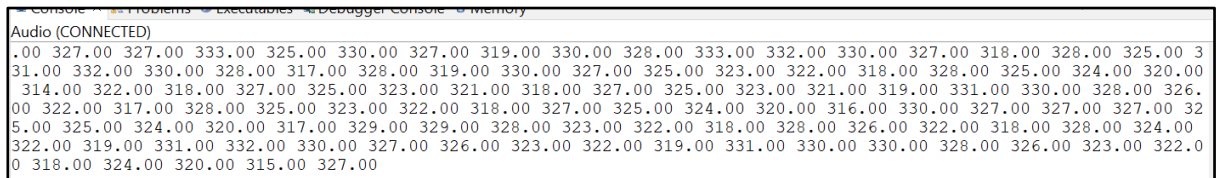
14. In the debug mode, go to the bottom right hand side corner, click on open console. Select the **Connection Type** as **Serial Port**, then click on **New**. In the new window, in **Connection name** give some name to your new connection, and select the **Serial port** correctly. Then click on **Finish** and then **Ok**. A console with the given name will be opened at the bottom of your screen.








15. Click on the **Resume** icon  to run your code. You should be able to see the value of audio sensor in the form of buffer containing 256 samples.



16. Before moving out of the debugging mode, click on **Disconnect** and close the console then click on the **Terminate** icon . You will be moved out of the debugging mode.



**Note:** All important steps and parts are highlighted with a red colour box for the proper understanding of the user. This document is for the use of education purpose only.