

## **Procedure to create project using Code Composer Studio V5.3(Real-time input – From Signal Generator or Microphone)**

*Create a New-Folder for your project inside CCS Folder*

1. Double click on **CCStudio** Icon.
2. Click on **Browse** and select proper location to save the project.
3. Click **ok**.

*Make sure that **Project Explorer Window** is visible if not click on **Restore Icon***

4. Click on **Project**→Select **New CCS project**.
5. Type **Project\_Name**.
6. Family **C6000**.
7. Variant →**DSK6713**(Select from 2<sup>nd</sup> Box).
8. Connection →**Spectrum Digital DSK-EVM -eZdsp on board USB Emulator**→Click **Finish**
9. Right Click on **Project\_Name**→Select **Properties**→go to **Build**→Expand **C6000 Compiler**
10. Select **Processor Options**→In **Target processor version**→ type “**6713**”
11. Expand **C6000 Linker**→Click on **File Search Path**.
12. Click on +(Include Library File)→Click on **File system**→ select **cs16713.lib**→Click **Open**→  
**ok**
13. Click on +(Include Library File)→Click on **File system**→ select **dsk6713bsl.lib**→Click  
**Open**→ **ok**
14. Click **Ok**
15. Right click on **Project\_Name**→ Select **add files**→go to **C-drive**→
16. Open **dsk6713** folder→Open **include** folder→ select **dsk6713.h**and **dsk6713\_aic23.h**→  
Click **Open**
17. Click **Ok** (Let the selection be as “**Copy Files**” which is default)
18. Click on **File**→ Click on **New**→Select **DSP/BIOS v5.x Configuration File**

*Make sure that your **Project\_Name** is displayed under **Project** option, if not click on **Browse** and select you **Project***

19. Click **Next** → Type **6713** in **Filter Platforms**→select **ti.platforms.dsk6713** → Click **Finish**

**Click on ok/yes on the windows which pop-up after this step even if there is error window**

## **Writing code:**

### **1. Type :**

```
#define CHIP_6713 1
#include "dsk6713.h"          // Header file for initialize function
#include "dsk6713_aic23.h"    // Header file for other functions

void main()
{
    // 1. Initialize Kit

    // 2. Open CODEC

    // 3. Set Sampling Frequency

    while(1)
    {
        // 4. Read Input

        // 5. Write Output

    } // End of While Loop

} // End of Main Function
```

**2. Double click on dsk6713.h and copy “*dsk6713\_init();*” and paste it below**  
*//1. Initialize kit in the main program. **Note:** DO NOT COPY “void”.*

**3. Double click on dsk6713\_aic23.h and copy “*DSK6713\_AIC23\_CodecHandle DSK6713\_AIC23\_openCodec(intid,DSK6713\_AIC23\_Config \*Config);*” and paste it below**  
*//2. Open CODEC in the main program.*

**4. Double click on dsk6713\_aic23.h and copy “*DSK6713\_AIC23\_setFreq(DSK6713\_AIC23\_CodecHandle hCodec, Uint32 freq);*” and paste it below**  
*//3. Set Sampling Frequency in the main program. **Note:** DO NOT COPY “void”.*

**5. Double click on dsk6713\_aic23.h and copy “*DSK6713\_AIC23\_read(DSK6713\_AIC23\_CodecHandle hCodec, Uint32 \*val)*” and paste it below**  
*//4. Read Input in the main program. **Note:** DO NOT COPY “Int16”.*

**6. Double click on dsk6713\_aic23.h and copy**

*“DSK6713\_AIC23\_write(DSK6713\_AIC23\_CodecHandle hCodec, Uint32 val);” and paste it below //5. Write Output in the main program. Note: DO NOT COPY “Int16”.*

**7. Copy “DSK6713\_AIC23\_CodecHandle” (a data type) and paste it before void main and after leaving space type “h” (a variable of data type DSK6713\_AIC23\_CodecHandle) .i.e., “DSK6713\_AIC23\_CodecHandle h”.**

**8. Copy “Uint32” (a data type) and paste it before void main and after leaving space, type “val” (a variable of data type Uint32). i.e., “Uint32 val”.**

**9. Copy “DSK6713\_AIC23\_Config” (a data type) and paste it before void main and after leaving space type “c” (a variable of data type DSK6713\_AIC23\_Config) .i.e., “DSK6713\_AIC23\_Config c”.**

**10. Double click on dsk6713\_aic23.h copy default config from opening brace ‘{’ till closing brace ‘}’ values and paste it as shown:**

```
“DSK6713_AIC23_Config c = { \
0x0017, /* Set-Up Reg 0    Left line input channel volume control */ \
/* LRS   0    simultaneous left/right volume: disabled */ \
/* LIM   0    left line input mute: disabled */ \
/* XX    00    reserved */ \
/* LIV   10111 left line input volume: 0 dB */ \
\
0x0017, /* Set-Up Reg 1    Right line input channel volume control */ \
/* RLS   0    simultaneous right/left volume: disabled */ \
/* RIM   0    right line input mute: disabled */ \
/* XX    00    reserved */ \
/* RIV   10111 right line input volume: 0 dB */ \
\
0x01f9, /* Set-Up Reg 2    Left channel headphone volume control */ \
/* LRS   1    simultaneous left/right volume: enabled */ \
/* LZC   1    left channel zero-cross detect: enabled */ \
/* LHV   1111001 left headphone volume: 0 dB */ \
\
0x01f9, /* Set-Up Reg 3    Right channel headphone volume control */ \
/* RLS   1    simultaneous right/left volume: enabled */ \
/* RZC   1    right channel zero-cross detect: enabled */ \
/* RHV   1111001 right headphone volume: 0 dB */ \
\
0x0011, /* Set-Up Reg 4    Analog audio path control */ \
/* X     0    reserved */ \
/* STA   00    sidetone attenuation: -6 dB */ \
/* STE   0    sidetone: disabled */ \
/* DAC   1    DAC: selected */ \
/* BYP   0    bypass: off */ \
/* INSEL 0    input select for ADC: line */ \
/* MICM   0    microphone mute: disabled */ \
\
```

```

        /* MICB 1 microphone boost: enabled */ \
        \
0x0000, /* Set-Up Reg 5 Digital audio path control */ \
        /* XXXXX 00000 reserved */ \
        /* DACM 0 DAC soft mute: disabled */ \
        /* DEEMP 00 deemphasis control: disabled */ \
        /* ADCHP 0 ADC high-pass filter: disabled */ \
        \
0x0000, /* Set-Up Reg 6 Power down control */ \
        /* X 0 reserved */ \
        /* OFF 0 device power: on (i.e. not off) */ \
        /* CLK 0 clock: on */ \
        /* OSC 0 oscillator: on */ \
        /* OUT 0 outputs: on */ \
        /* DAC 0 DAC: on */ \
        /* ADC 0 ADC: on */ \
        /* MIC 0 microphone: on */ \
        /* LINE 0 line input: on */ \
        \
0x0043, /* Set-Up Reg 7 Digital audio interface format */ \
        /* XX 00 reserved */ \
        /* MS 1 master/slave mode: master */ \
        /* LRSWAP 0 DAC left/right swap: disabled */ \
        /* LRP 0 DAC lrp: MSB on 1st BCLK */ \
        /* IWL 00 input bit length: 16 bit */ \
        /* FOR 11 data format: DSP format */ \
        \
0x0081, /* Set-Up Reg 8 Sample rate control */ \
        /* X 0 reserved */ \
        /* CLKOUT 1 clock output divider: 2 (MCLK/2) */ \
        /* CLKIN 0 clock input divider: 2 (MCLK/2) */ \
        /* SR,BOSR 00000 sampling rate: ADC 48 kHz DAC 48 kHz */ \
        /* USB/N 1 clock mode select (USB/normal): USB */ \
0x0001 /* Set-Up Reg 9 Digital interface activation */ \
        /* XX..X 00000000 reserved */ \
        /* ACT 1 active */ \
};

```

**Note:** semicolon ‘;’ should be there after closing brace ‘}’ indicating termination of line.

## 11. In the main program

- i) Replace “0x0011” with “0x0015”.
- ii) Replace “*DSK6713\_AIC23\_CodecHandle hCodec*” with “*h*”.
- iii) Replace “*DSK6713\_AIC23\_Config \*Config*” with “*&c*”.
- iv) Replace “*UInt32 \*val*” with “*&val*”.
- v) Replace “*UInt32 val*” with “*val*”.
- vi) Replace “*UInt32 freq*” with required frequency values. Refer */\*Frequency Definitions\*/* in *dsk6713\_aic23.h*.
- vii) Replace “*DSK6713\_AIC23\_CodecHandle DSK6713\_AIC23\_openCodec(int id, DSK6713\_AIC23\_Config \*Config);*” with “*h=DSK6713\_AIC23\_openCodec(0, &c)*”.

## 12. Once completed the entire code should look as below

```
#define CHIP_6713 1
#include "dsk6713.h" // Header file for initialize function
#include "dsk6713_aic23.h" // Header file for other functions

DSK6713_AIC23_CodecHandle h;
DSK6713_AIC23_Config c = { \
    0x0017, /* Set-Up Reg 0 Left line input channel volume control */ \
        /* LRS 0 simultaneous left/right volume: disabled */ \
        /* LIM 0 left line input mute: disabled */ \
        /* XX 00 reserved */ \
        /* LIV 10111 left line input volume: 0 dB */ \
        \
    0x0017, /* Set-Up Reg 1 Right line input channel volume control */ \
        /* RLS 0 simultaneous right/left volume: disabled */ \
        /* RIM 0 right line input mute: disabled */ \
        /* XX 00 reserved */ \
        /* RIV 10111 right line input volume: 0 dB */ \
        \
    0x01f9, /* Set-Up Reg 2 Left channel headphone volume control */ \
        /* LRS 1 simultaneous left/right volume: enabled */ \
        /* LZC 1 left channel zero-cross detect: enabled */ \
        /* LHV 1111001 left headphone volume: 0 dB */ \
        \
    0x01f9, /* Set-Up Reg 3 Right channel headphone volume control */ \
        /* RLS 1 simultaneous right/left volume: enabled */ \
        /* RZC 1 right channel zero-cross detect: enabled */ \
        /* RHV 1111001 right headphone volume: 0 dB */ \
        \
    0x0015, /* Set-Up Reg 4 Analog audio path control */ \
        /* X 0 reserved */ \
        /* STA 00 sidetone attenuation: -6 dB */ \
        /* STE 0 sidetone: disabled */ \
        /* DAC 1 DAC: selected */ \
        /* BYP 0 bypass: off */ \
        /* INSEL 0 input select for ADC: line */ \
        /* MICM 0 microphone mute: disabled */ \
        /* MICB 1 microphone boost: enabled */ \
        \
    0x0000, /* Set-Up Reg 5 Digital audio path control */ \
        /* XXXXX 00000 reserved */ \
        /* DACM 0 DAC soft mute: disabled */ \
        /* DEEMP 00 deemphasis control: disabled */ \
        /* ADCHP 0 ADC high-pass filter: disabled */ \
        \
    0x0000, /* Set-Up Reg 6 Power down control */ \
        /* X 0 reserved */ \
        /* OFF 0 device power: on (i.e. not off) */ \
        /* CLK 0 clock: on */ \
        /* OSC 0 oscillator: on */ \
        /* OUT 0 outputs: on */ \
        /* DAC 0 DAC: on */ \
        /* ADC 0 ADC: on */ \
        /* MIC 0 microphone: on \

```

```

        /* LINE 0 line input: on */
0x0043, /* Set-Up Reg 7 Digital audio interface format */
        /* XX 00 reserved */
        /* MS 1 master/slave mode: master */
        /* LRSWAP 0 DAC left/right swap: disabled */
        /* LRP 0 DAC lrp: MSB on 1st BCLK */
        /* IWL 00 input bit length: 16 bit */
        /* FOR 11 data format: DSP format */
0x0081, /* Set-Up Reg 8 Sample rate control */
        /* X 0 reserved */
        /* CLKOUT 1 clock output divider: 2 (MCLK/2) */
        /* CLKIN 0 clock input divider: 2 (MCLK/2) */
        /* SR,BOSR 000000 sampling rate: ADC 48 kHz DAC 48 kHz */
        /* USB/N 1 clock mode select (USB/normal): USB */
0x0001 /* Set-Up Reg 9 Digital interface activation */
        /* XX..X 00000000 reserved */
        /* ACT 1 active */
};

Uint32 val;

void main()
{
    // 1. Initialize Kit
    DSK6713_init();

    // 2. Open CODEC
    h = DSK6713_AIC23_openCodec(0, &c);

    // 3. Set Sampling Frequency
    DSK6713_AIC23_setFreq(h, 1); // value '1' for 8kHz Sampling Rate

    while(1) // To read values continuously
    {
        // 4. Read Input
        DSK6713_AIC23_read(h, &val);

        // 5. Write Output
        DSK6713_AIC23_write(h, val);

    } // End of While Loop

} // End of Main Function

```

### **To run the program:**

1. Right click on **Project Name**→Click on **Build Project**  
*Check for errors in “**Problems**” window. Rectify the errors if any and build once again.*  
*Don't proceed until you have **zero errors**, you can ignore warnings*
2. Click on **View**→ Select **Target Configuration**→Expand **User Defined**
3. Right click on **C6713DSK.ccxml**→Select **Launch Selected Configuration**  
*CC Studio will migrate from **Edit Perspective** to **Debug Perspective***
4. Click on **Run**→Select **Connect Target**
5. Click on **Run**→Select **Load**→ Select **Load Program**→Click on **Browse Project**
6. Expand **Project\_Name**→Expand **Debug** →Select **Project\_Name.out** file→Click **Ok**→**Ok**
7. Click on **Run**→ Select **Resume**.