Digital Signal Processing and Applications with the TMS320C6713 DSK

Day 1

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Texas Instruments Technology for Innovators™

Workshop Goals

- Become familiar with
  - DSP basics
  - TMS320C6713 floating point DSP architecture
  - TMS320C6713 DSP starter kit (DSK)
  - Code composer studio integrated development environment (IDE)
  - Matlab design and analysis tools
- Learn how to program the C6713
  - Writing and compiling code
  - Fixing errors
  - Downloading code to the target and executing
  - Debugging
- Write and run useful programs on the C6713 DSK
- Learn about DSP applications
- Learn where to find help

Take Home Items

- "*Digital Signal Processing and Applications with the C6713 and C6416 DSK*" by Rulph Chassaing, 2005
- **Texas Instruments TMS320C6713 DSK** including
  - DSK board with TMS320C6713 DSP chip
  - USB cable
  - Power supply
  - CD with Code composer studio IDE (v3.1) and electronic documentation
  - DSK technical reference manual
  - DSK quick start installation guide
  - Matlab/Simulink trial CD and other promotional material

C6713 DSK Overview

- 225 MHz TMS320C6713 ***floating point*** DSP
- AIC23 stereo codec (ADC and DAC)
  - Ideal for audio applications
  - 8-96 kHz sample rates
- Memory
  - 16 MB dynamic RAM
  - 512 kB nonvolatile FLASH memory
- General purpose I/O
  - 4 LEDs
  - 4 DIP switches
- USB interface to PC

C6713 DSK
Functional Block Diagram

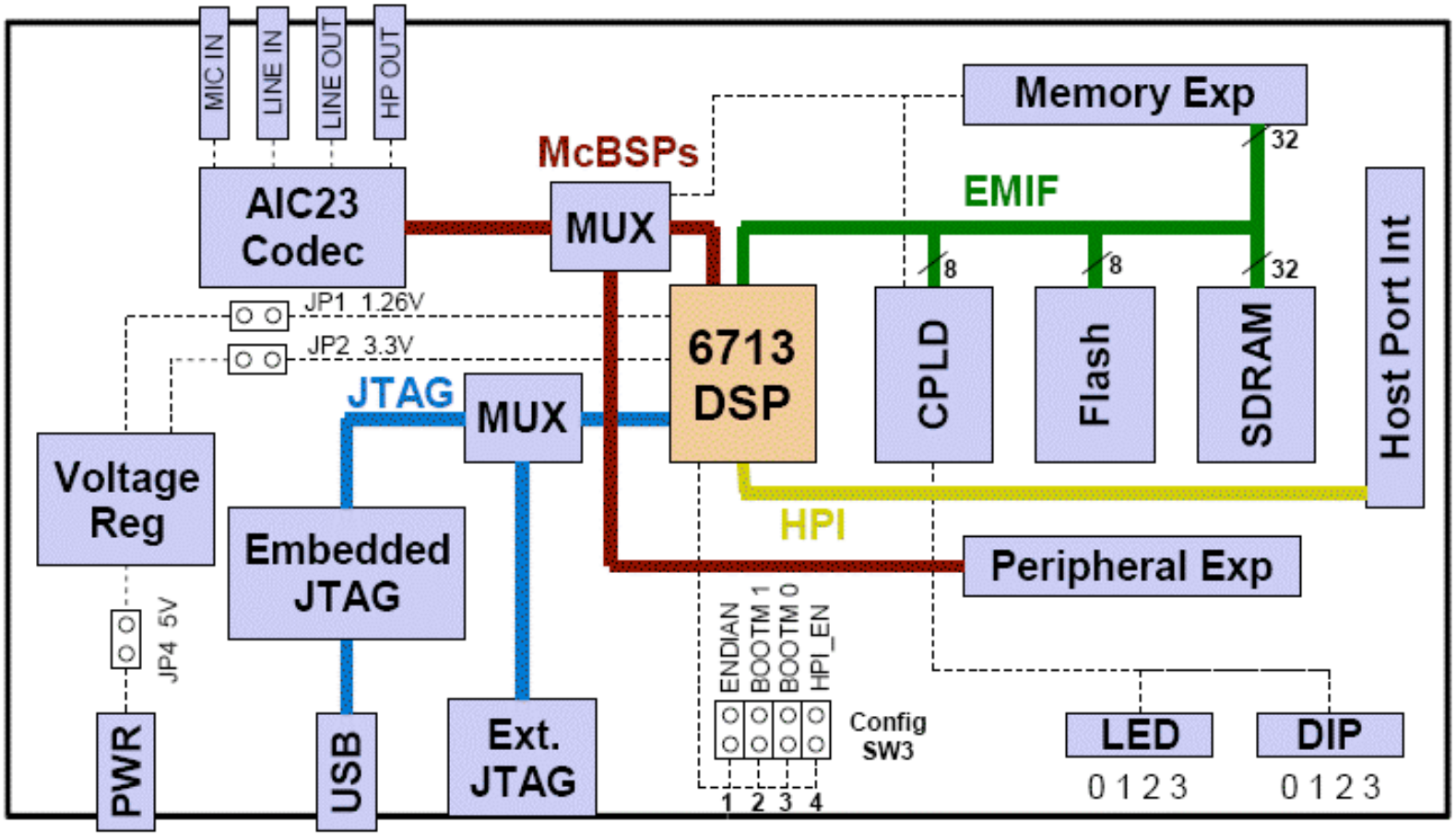
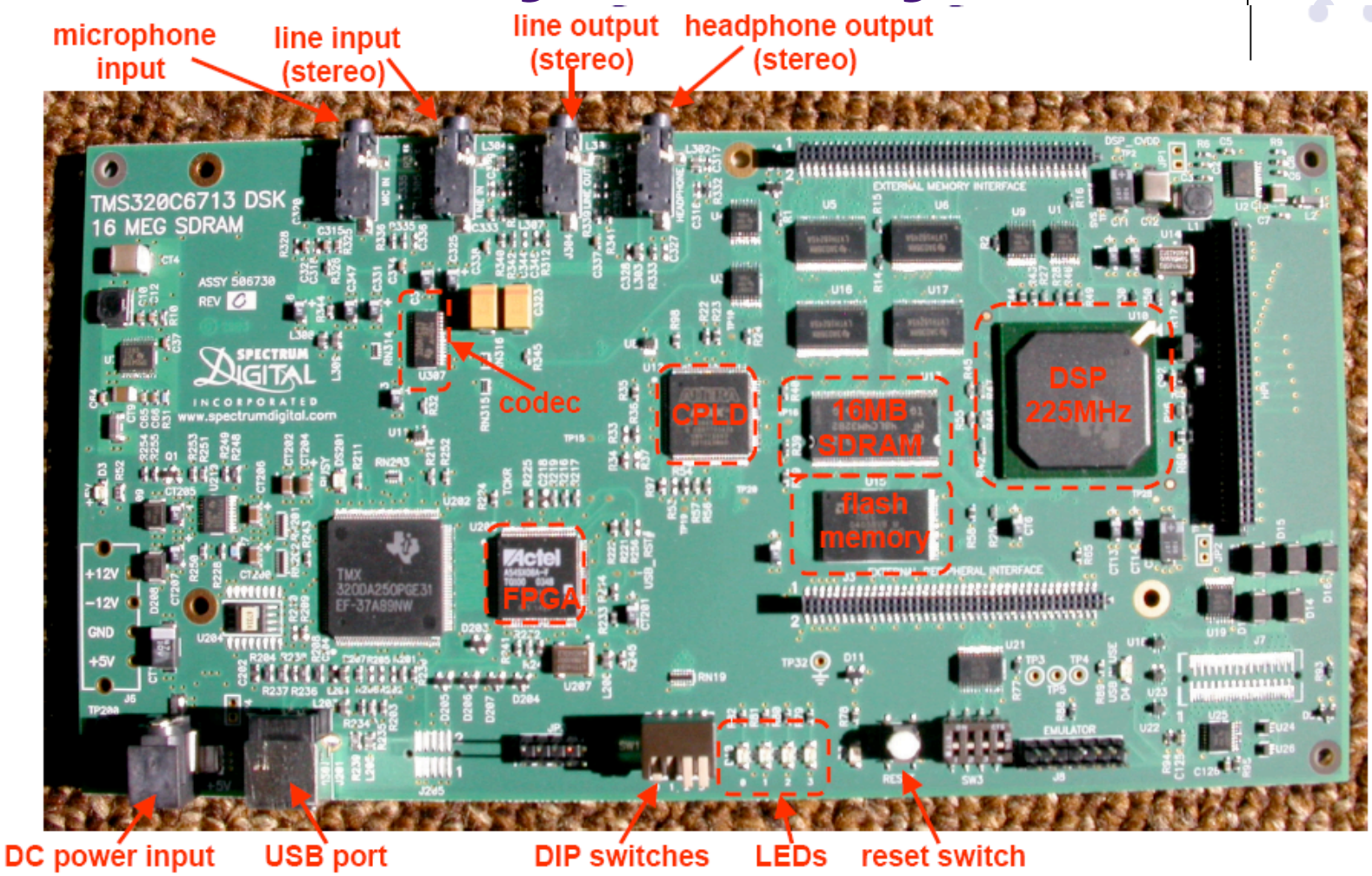


Figure 1-1, Block Diagram C6713 DSK

C6713 DSK Physical Layout

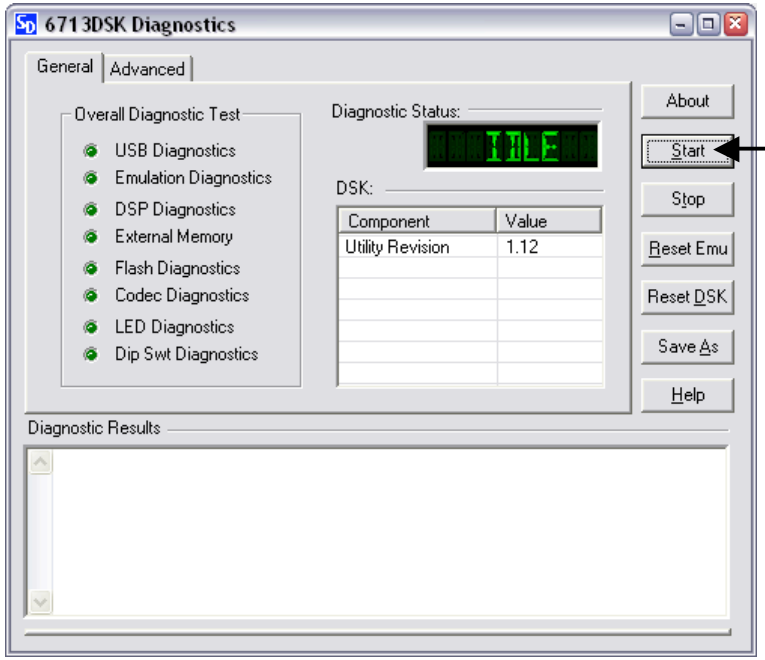


Is my DSK working? DSK Power On Self Test

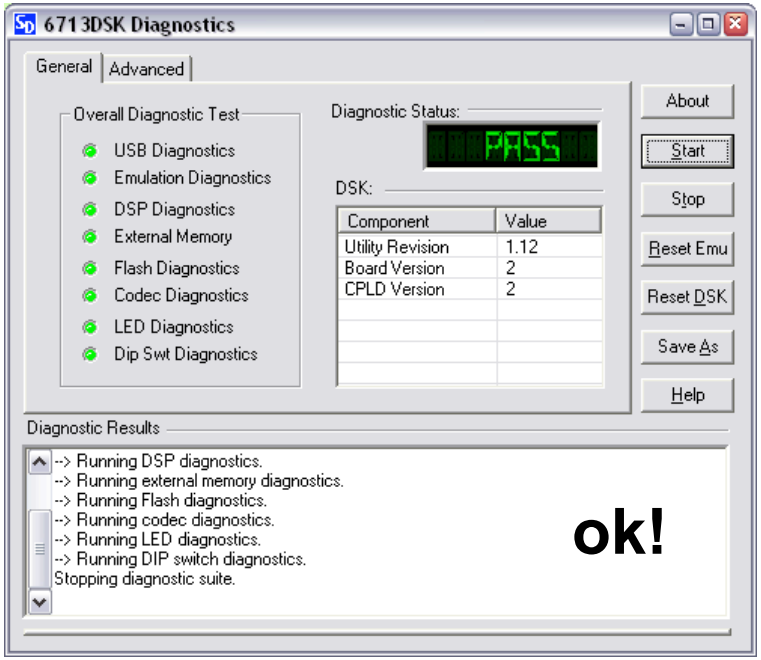
- Power up DSK and watch LEDs
- Power On Self Test (POST) program stored in FLASH memory automatically executes
- POST takes 10-15 seconds to complete
- All DSK subsystems are automatically tested
- During POST, a 1kHz sinusoid is output from the AIC23 codec for 1 second
  - Listen with headphones or watch on oscilloscope
- If POST is successful, all four LEDs blink 3 times and then remain on

Is my DSK working?
DSK Diagnostic Utility

- Install CCS 3.1
  - Directions in “Quick Start Installation Guide”
  - Diagnostic utility automatically installed



press start



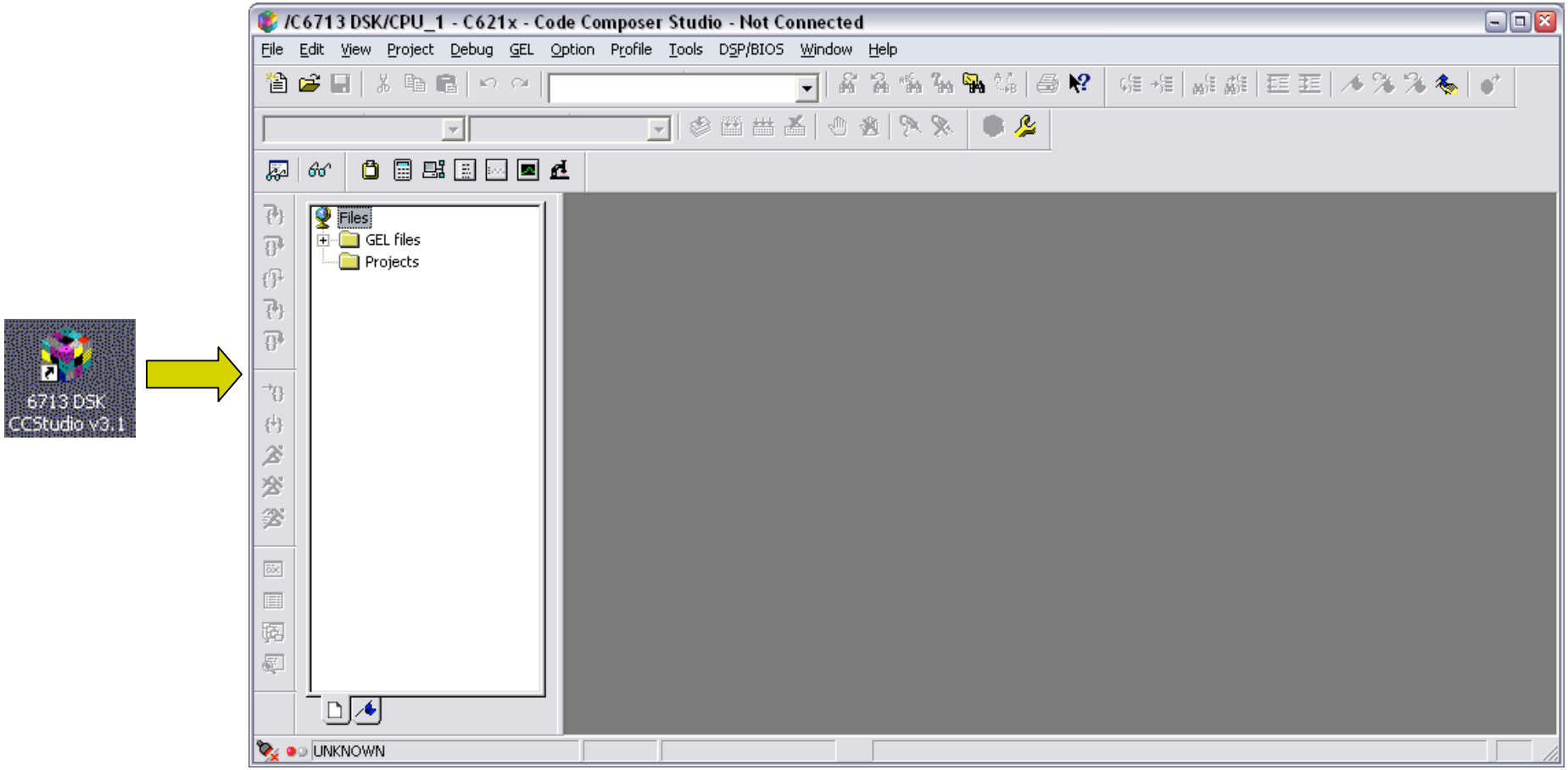
ok!

Code Composer Studio IDE

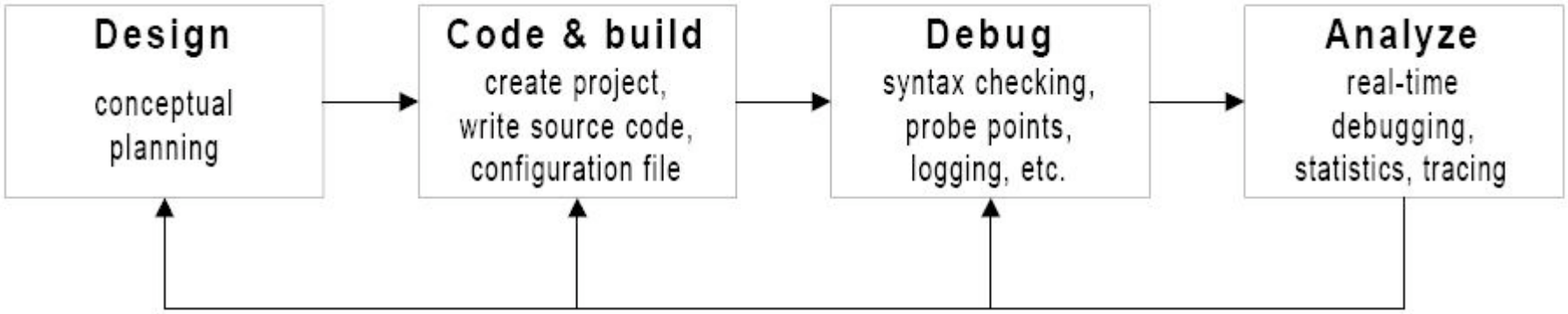
- Connect power supply to DSK
- Wait for POST to complete
- Connect USB cable from PC to DSK
  - If this is the first time connecting the DSK, you may be asked to install a driver. The driver is on the Code Composer Studio CD and will automatically be found by Windows if the CD is in the drive.
- Launch Code Composer Studio C6713 DSK →

- CCS will load and wait for your input

Code Composer Studio IDE



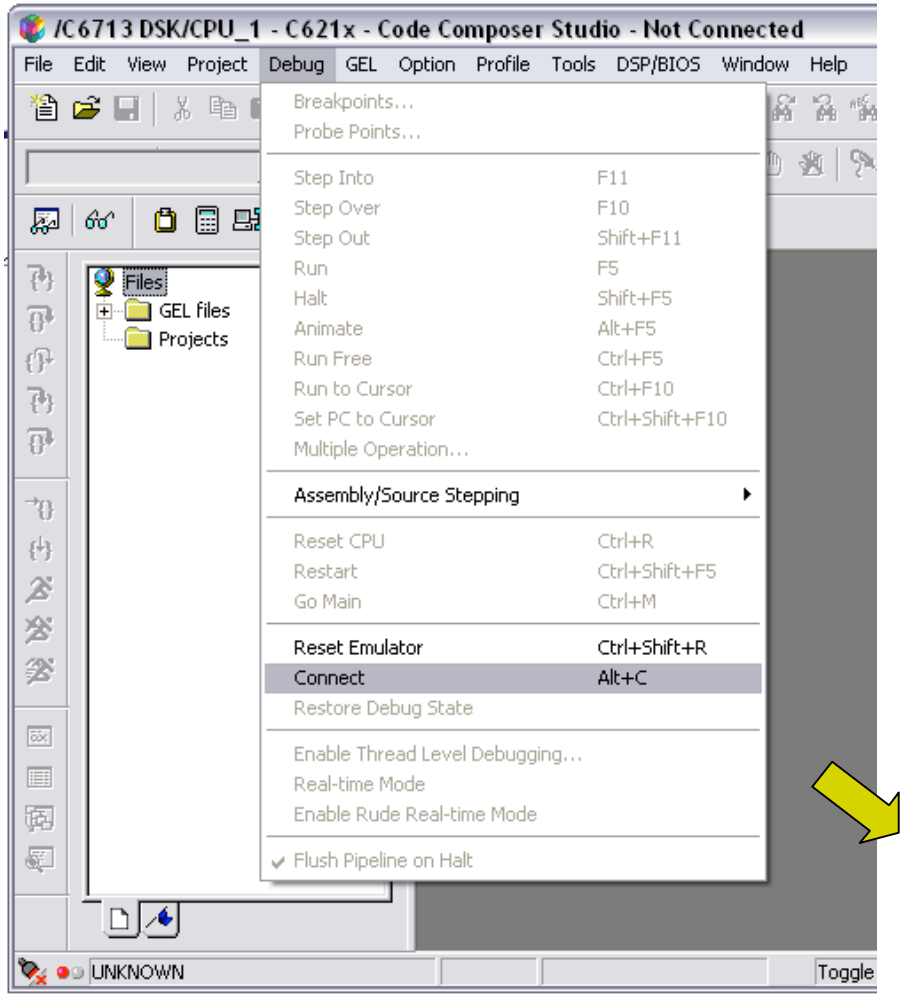
CCS Integrated Development Environment

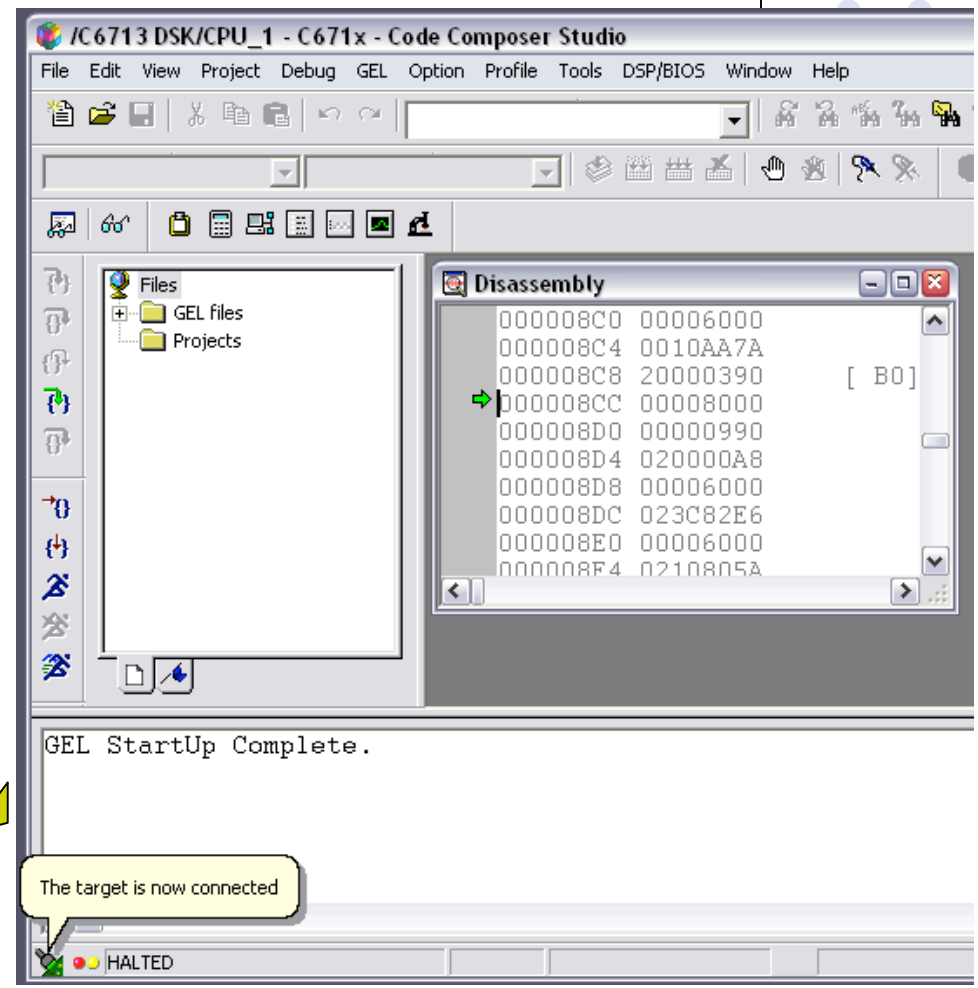


Useful TI documentation (available online or on your hard drive):
SPRU509F.PDF CCS v3.1 IDE Getting Started Guide
C6713DSK.HLP C6713 DSK specific help material

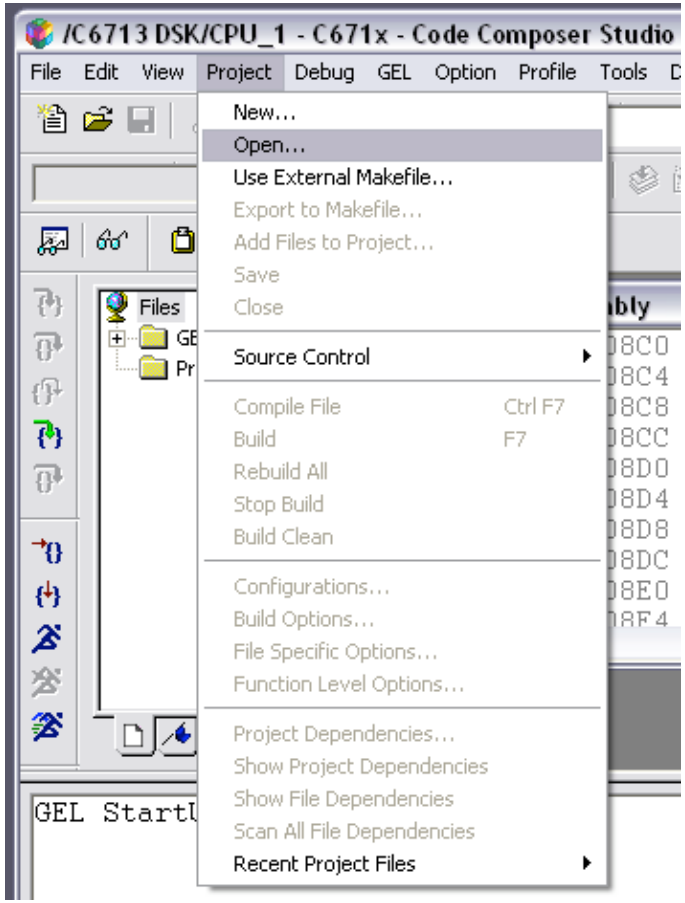
Note that your DSK includes CCS v3.1. Updates and patches are available after registering CCS.

Connecting to the C6713 DSK

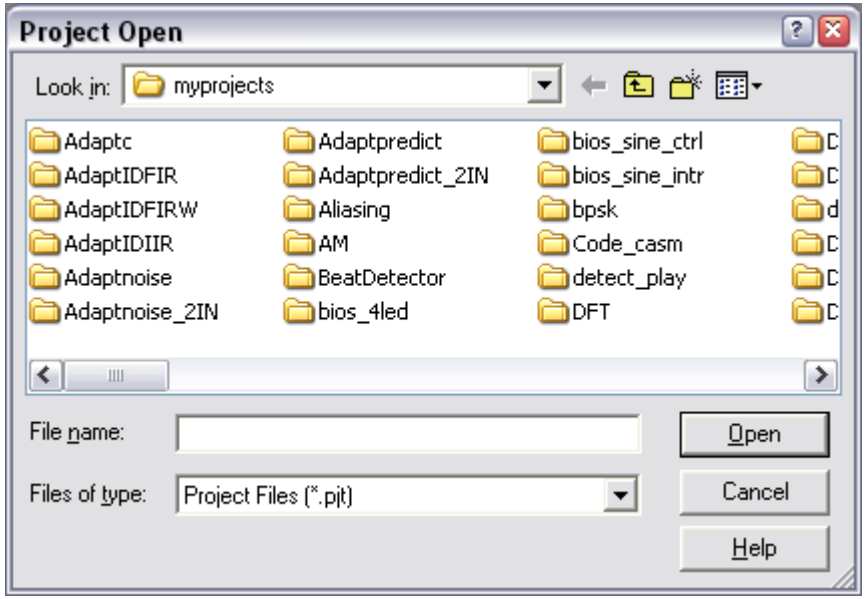




Opening an Existing Project



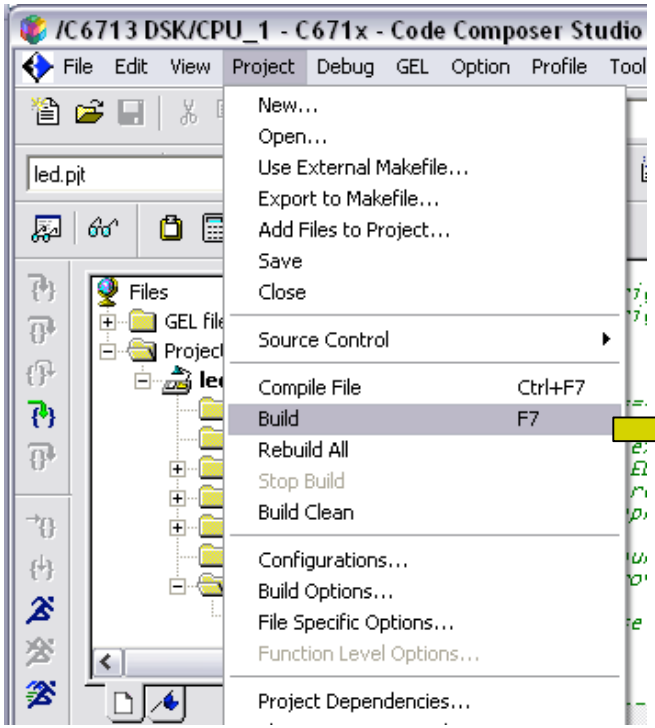
Project->Open

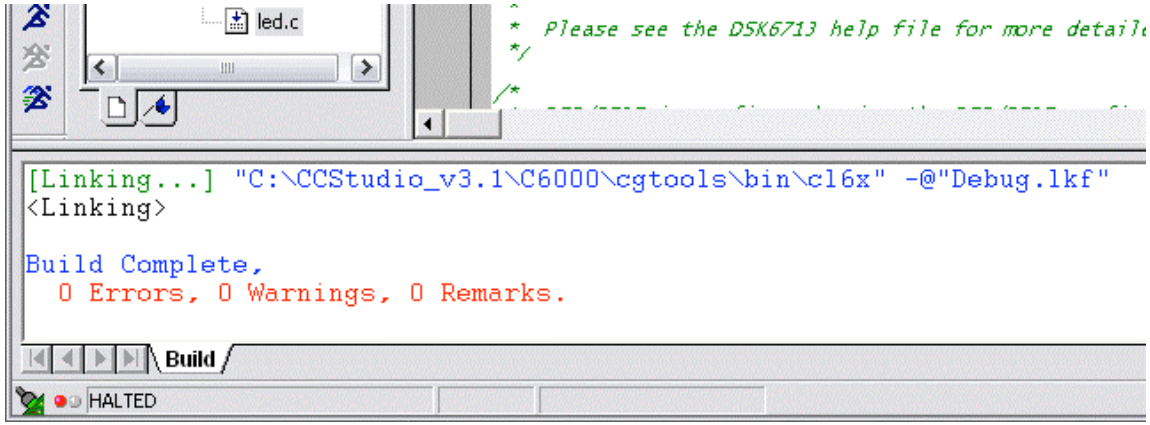


Select a .PJT file and press “Open”. Chassaing example projects should be in **c:\CCStudio_v3.1\myprojects**
Other example projects for the C6713 can be found in **c:\CCStudio_v3.1\examples\dsk6713**

Compiling/Building a Project

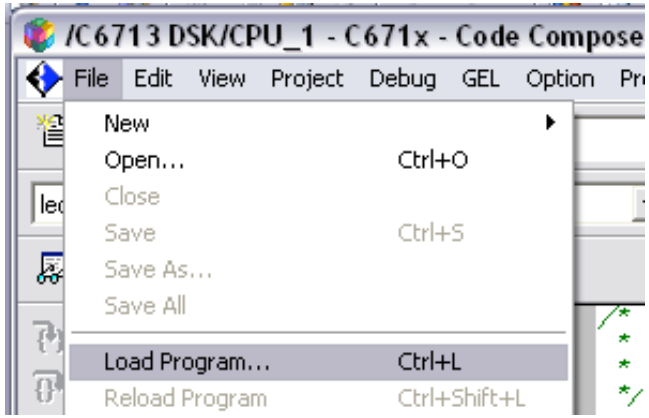
Project->Build (F7)

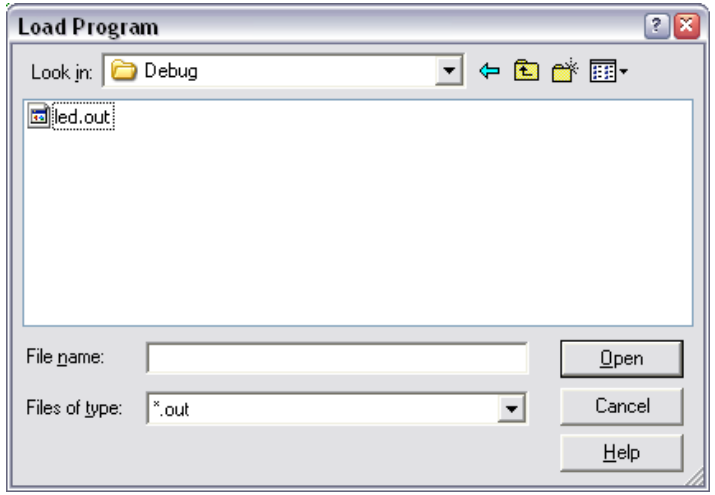




Loading and Running a Project on the C6713 DSK

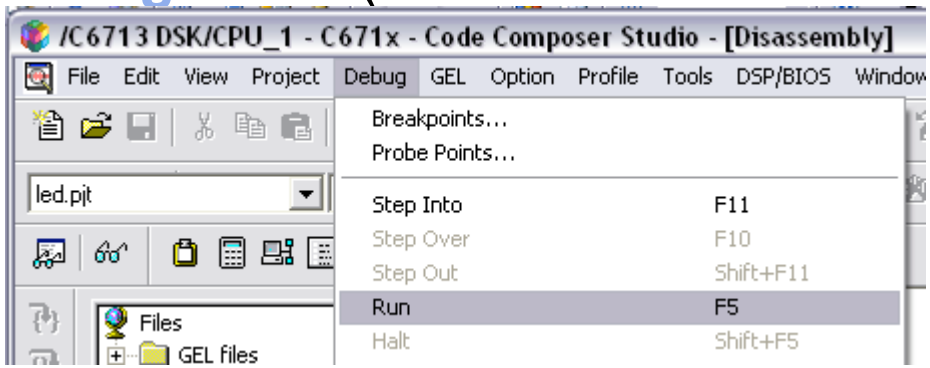
File-> Load Program (ctrl+L)





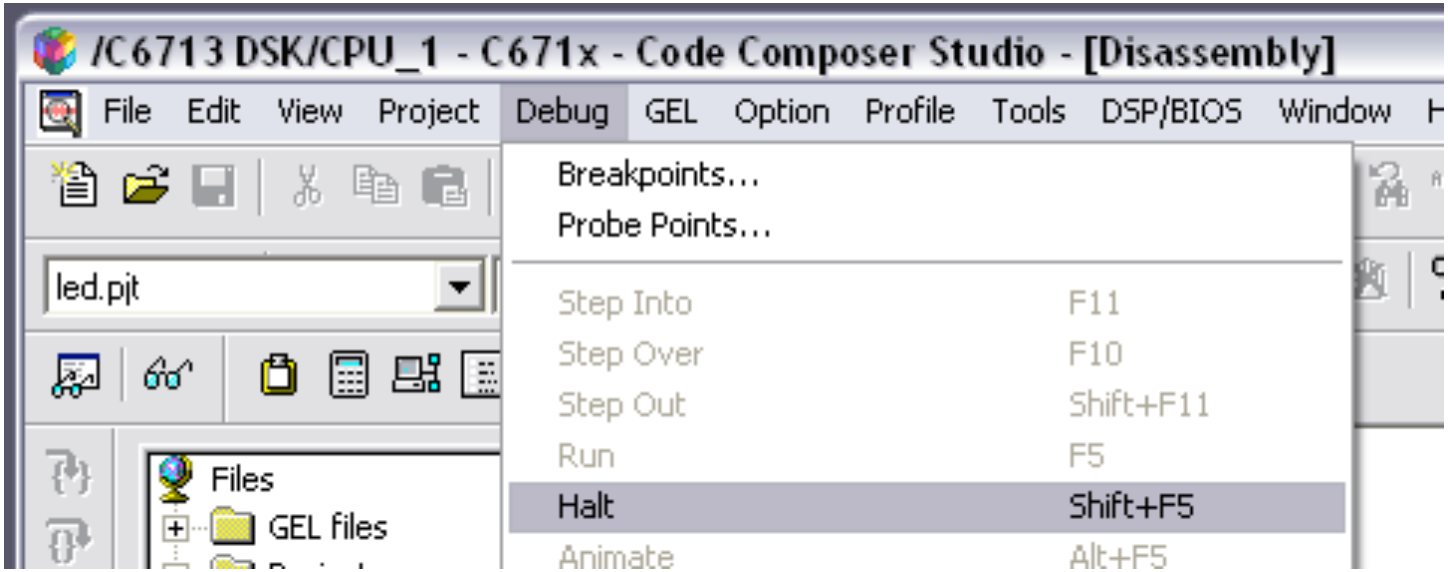
Select the .out file in the project\Debug directory. Program is sent to DSK.

Debug->Run (F5 or the Run button)



Halting a Running Program on the C6713 DSK

Debug->Halt (shift+F5 or the Halt button ).



Chassaing textbook examples: Fixing the search path

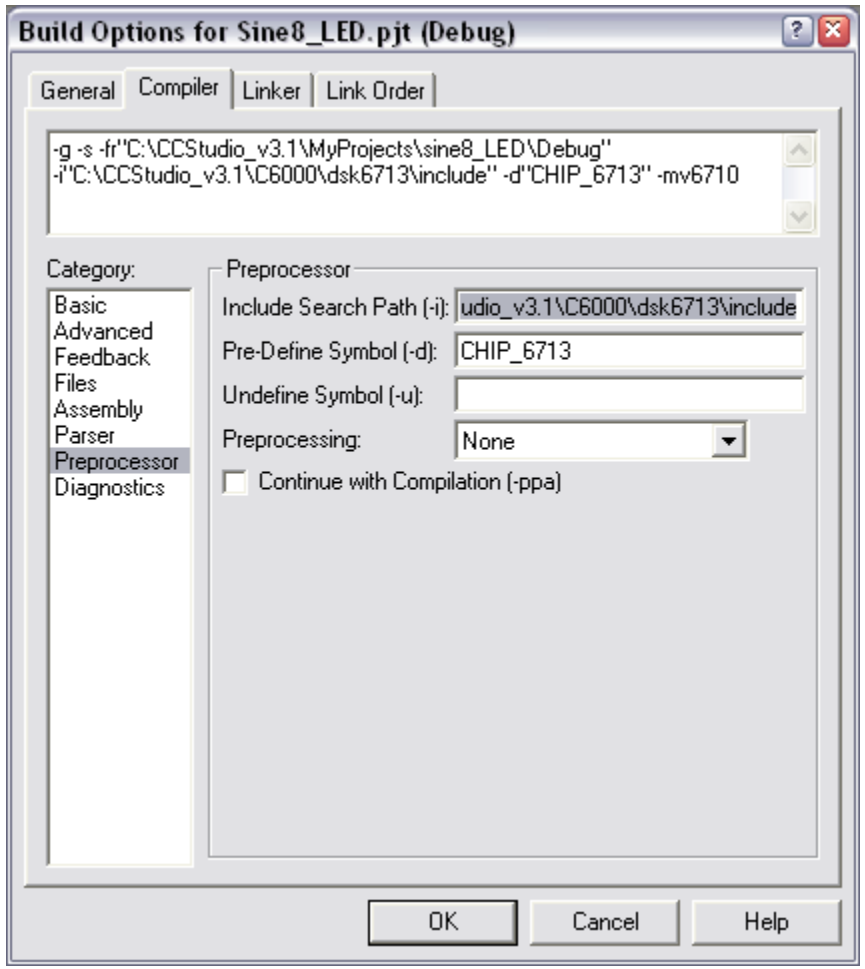
Add C:\CCStudio_v3.1\C6000\dsk6713\include to the search path

Project ->

Build Options ->

[Compiler tab] ->

[Preprocessor category]



Chassaing textbook examples: Fixing the mem model

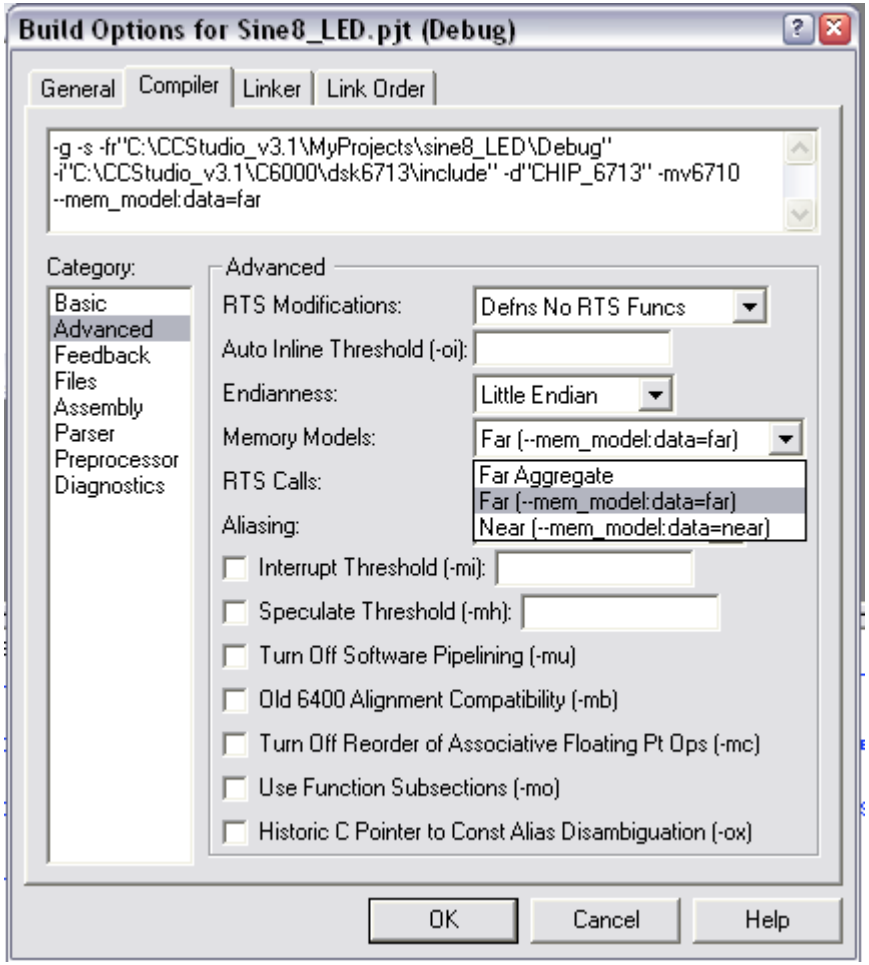
Change the memory model to “data=far”

Project ->

Build Options ->

[Compiler tab] ->

[Advanced category]

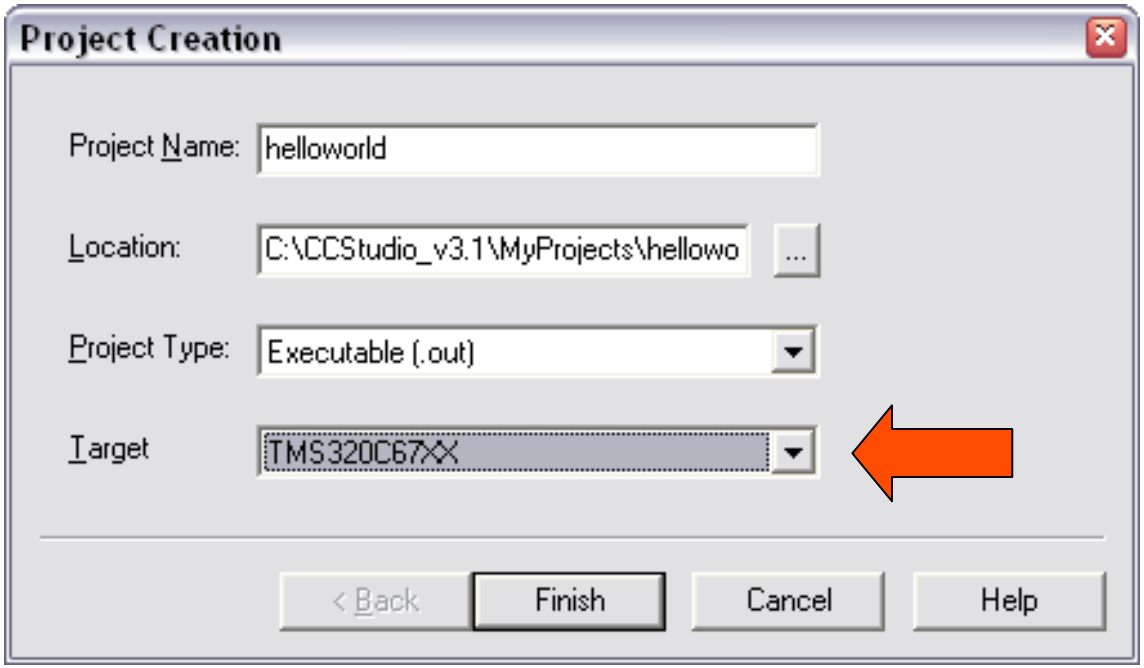


Things to Try

- Open Sin8_LED project and fix the search path and the memory model (see previous pages). Then build, load, and run it.
  - Press DIP switch 0. You should see LED 0 light up and a 1kHz sinusoid should appear on the left channel of the codec. This is a good test to see if the DSK is working.
- Make an error in the source code Sin8_LED.c and build the project to see what happens.
- Change the amplitude of the sinusoid (gain variable), rebuild, reload, and see what happens.
- Modify the code to generate a 500Hz sinusoid.
- Open, build, and load other projects in “myprojects”

Creating a New Project (1 of 5)

1. Create new project
   Project->New



Creating a New Project (2 of 5)

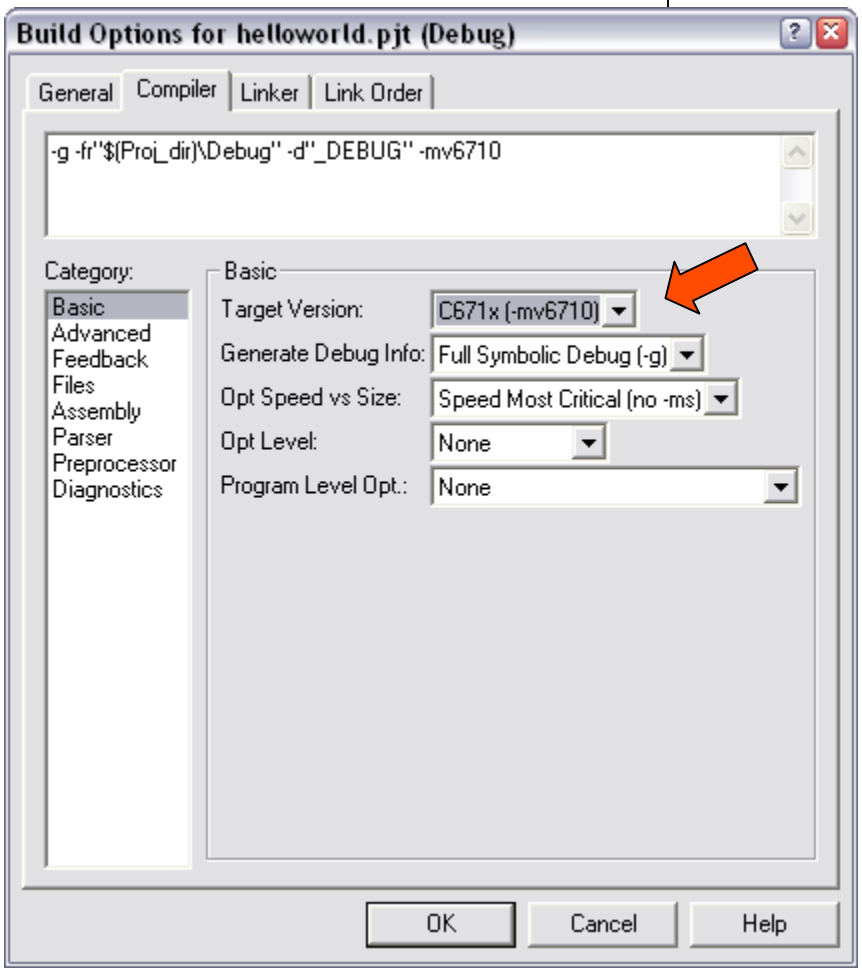
2. Write your C code:
   File->New->Source File
3. Save it in your project directory (make sure it has a .c extension):
   File->Save
4. Add your C code to the project:
   Project->Add Files to Project

Creating a new project (3 of 5)

5. Add required support files to project
   Project->Add Files to Project
   a) **myprojects\support\c6713dsk.cmd**
      [linker command file – this or another cmd file is **required**]
   b) **c6000\cgtools\lib\rts6700.lib**
      [run-time support library functions - **required**]
6. Add optional support files to project, e.g.
   Project->Add Files to Project
   a) **myprojects\support\vectors_poll.asm** or **vectors_intr.asm**
      [used to set up interrupt vectors]
   b) **c6000\dsk6713\lib\dsk6713bsl.lib**
      [DSK board support library functions – useful for interfacing to the codec, DIP switches, and LEDs]
   c) **c6000\bios\lib\csl6713.lib**
      [chip support library functions]

Creating a New Project (4 of 5)

7. Set up the build options for C6713:
   Project -> Build Options (compiler tab)
   - Make sure target version is C671x
   - Also make sure Opt(imization) Level is “none” - this will help with debugging



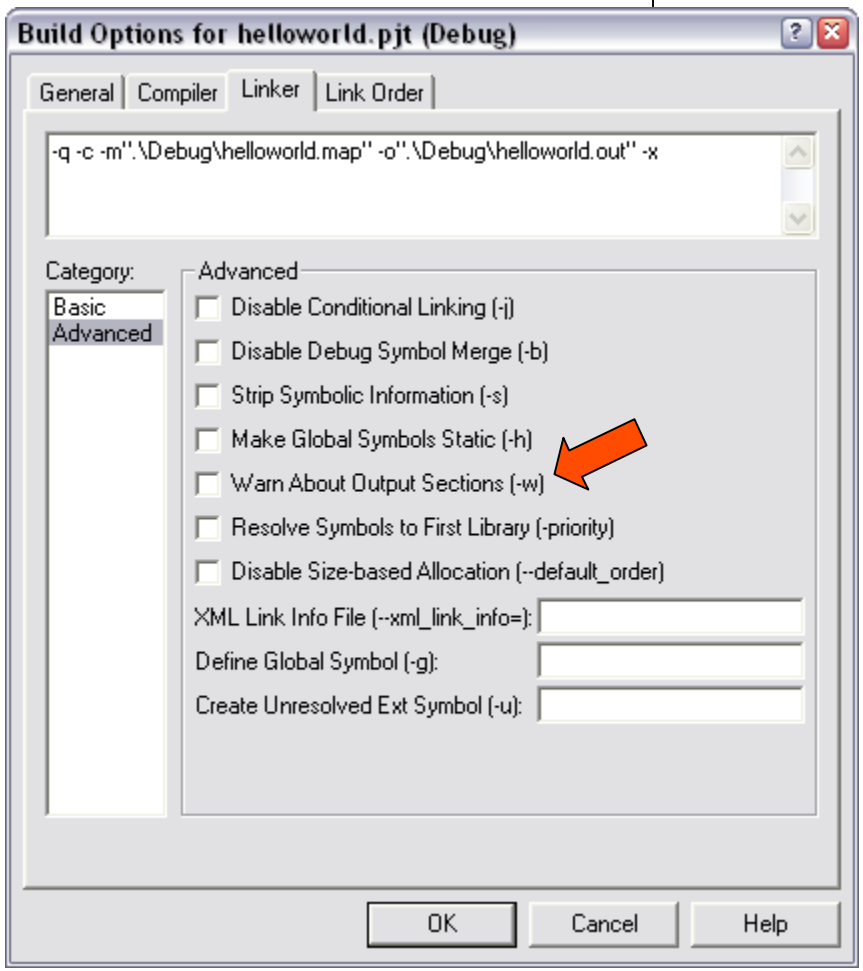
Creating a New Project (5 of 5)

8. Scan all file dependencies to automatically bring all header files and includes into the project:
   Project -> Scan all file dependencies
9. Build the project:
   Project -> Build
10. If successful, load the .out file to the DSK:
    File -> Load Program
    Select the Debug directory. Select the .out file.
11. Run it:
    Debug -> Run or F5 or the run button.

Optional: Suppress linker warnings

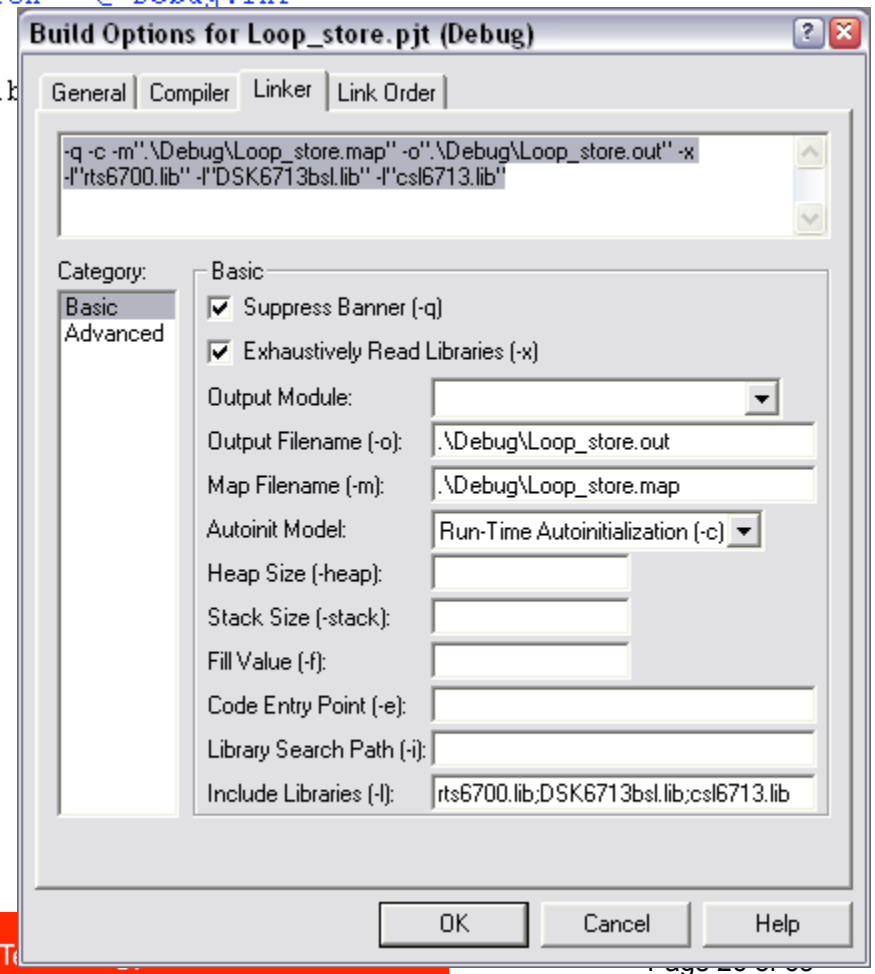
Project->Build Options
(linker tab)

Uncheck “warn about output sections” (or put in values for stack and heap in the Basic category)



Tip: Problems finding files during linking

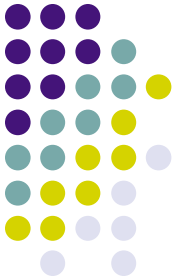
```
[Loop_store.c] "C:\CCStudio_v3.1\C6000\cgtools\bin\cl6x" -g -q -fr"C:/

[Linking...] "C:\CCStudio_v3.1\C6000\cgtools\bin\cl6x" -@"Debug.lkf"
<Linking>
>> C:\DOCUME~1\drb\LOCALS~1\Temp\TI5643, line 21:
              can't find input file 'DSK6713bsl.lib

>> Compilation failure

Build Complete.
```



Problem is caused by a bad path for the include libraries in the linker options (Project -> Build Options -> Linker tab)

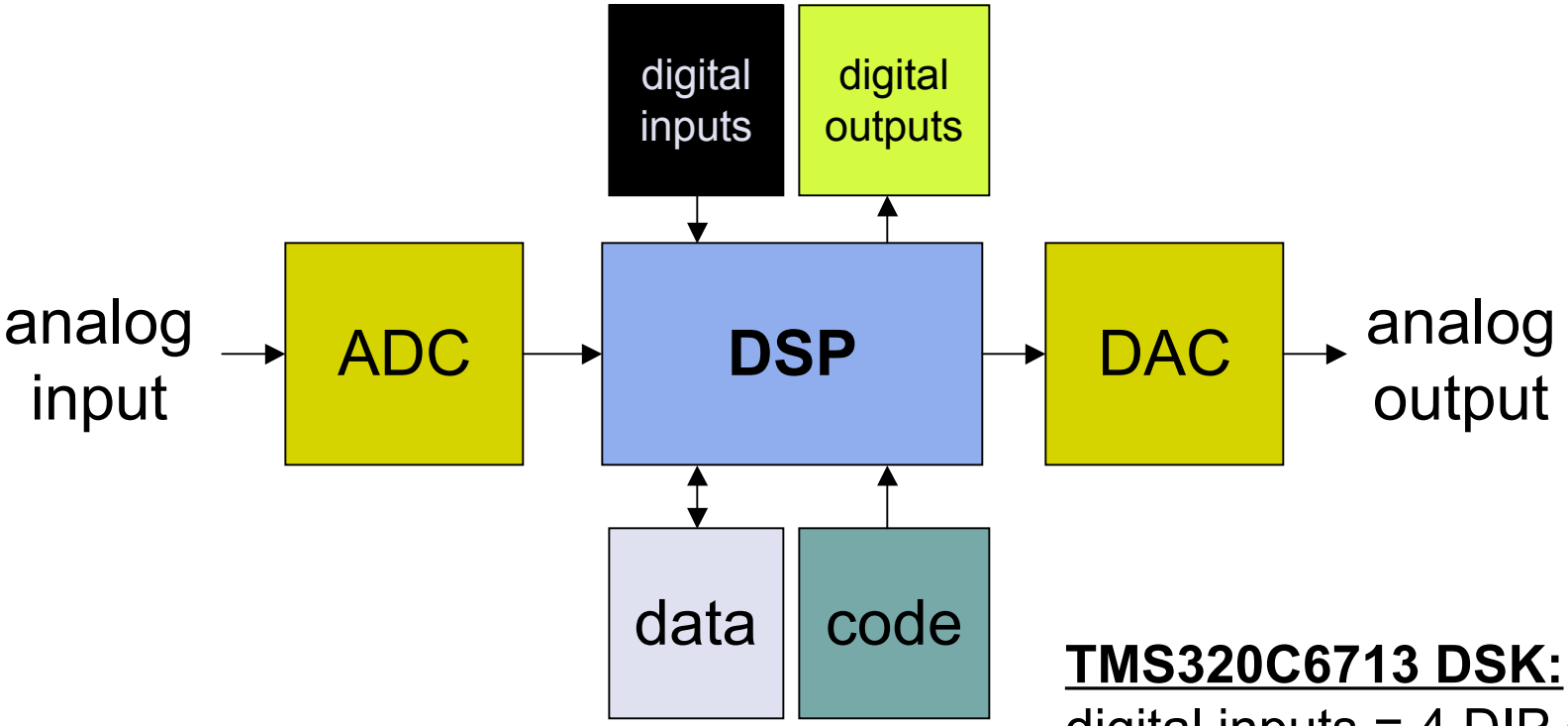
A fix for this is to remove rts6700.lib, DSK6713bsl.lib, and csl6713.lib from the linker options and add these files manually (Project -> Add files to Project…)

A Simple Program to Try: “helloworld”



```
// helloworld.c
// D. Richard Brown III
// 9-Oct-2006

#include <stdio.h>

void main()
{

        printf("Hello world.\n");

}
```

More Interesting Programs: Interfacing with the Real World



<u>TMS320C6713 DSK:</u>
digital inputs = 4 DIP switches
digital outputs = 4 LEDs
ADC and DAC = AIC23 codec

Interfacing with the DIP Switches and LEDs

LED and DIP switch interface functions are provided in **dsk6713bsl.lib**.

- Initialize DIP/LEDs with
  DSK6713_DIP_init() and/or DSK6713_LED_init()
- Read state of DIP switches with
  DSK6713_DIP_get(n)
- Change state of LEDs with
  DSK6713_LED_on(n) or
  DSK6713_LED_off(n) or
  DSK6713_LED_toggle(n)

where n=0, 1, 2, or 3.

Documentation is available in
C:\CCStudio_v3.1\docs\hlp\c6713dsk.hlp

Interfacing with the AIC23 codec: C6x Interrupt Basics

- Interrupt sources must be mapped to interrupt events
  - 16 “interrupt sources” (timers, serial ports, …)
  - 12 “interrupt events” (INT4 to INT15)
- Interrupt events have associated “interrupt vectors”. An “interrupt vector” is a special pointer to the start of the “interrupt service routine” (ISR).
- Interrupt vectors must be set up in your code (usually in the file “vectors.asm”).
- You are also responsible for writing the ISR.

Setting up an interface with the AIC23 Codec (step 1 of 3)

We can write the ISR first:

```
interrupt void serialPortRcvISR()
{
    Uint32 temp;

    temp = MCBSP_read(DSK6713_AIC23_DATAHANDLE);  // read L+R channels
    MCBSP_write(DSK6713_AIC23_DATAHANDLE,temp);   // write L+R channels
}
```

Remarks:

- **MCBSP_read()** requests samples from the codec's ADC
- **MCBSP_write()** sends samples to the codec's DAC
- This ISR simply reads in samples and then sends them back out.

Codec data format and how to separating the left/right channels

```
// we can use the union construct in C to have
// the same memory referenced by two different variables
union {Uint32 combo; short channel[2];} temp;
```


```
// the McBSP functions require that we
// read/write data to/from the Uint32 variable
temp.combo = MCBSP_read(DSK6713_AIC23_DATAHANDLE);
MCBSP_write(DSK6713_AIC23_DATAHANDLE, temp.combo);

// but if we want to access the left/right channels individually
// we can do this through the short variables
Leftchannel = temp.channel[1];
Rightchannel = temp.channel[0];
```

Setting up an interface with the AIC23 Codec (step 2 of 3)

- Now we can set up the interrupt vector to point to the ISR.
- In this example, our ISR is called "**serialPortRcvISR**".
- We will link the codec interrupt event to **INT15**.
- Here is the appropriate code in the **vectors.asm** file:

```
INT15:
    MVKL .S2 _serialPortRcvISR, B0
    MVKH .S2 _serialPortRcvISR, B0
    B    .S2 B0
    NOP
    NOP
    NOP
    NOP
    NOP
```

Setting up an interface with the AIC23 Codec (step 3 of 3)

Initialization steps:

1. Initialize the DSK
2. Open the codec with the default configuration.
3. Configure multi-channel buffered serial port (McBSP)
4. Configure codec parameters, e.g. set the sampling rate
5. Configure and enable interrupts
6. Do normal processing (we just enter a loop here)

```c
interrupt void serialPortRcvISR(void);                    // ISR function prototype

void main()
{
    DSK6713_init();     // Initialize the board support library, must be called first
    hCodec = DSK6713_AIC23_openCodec(0, &config);          // Open the codec

    // Configure buffered serial ports for 32 bit operation
    // This allows transfer of both right and left channels in one read/write
    MCBSP_FSETS(SPCR1, RINTM, FRM);
    MCBSP_FSETS(SPCR1, XINTM, FRM);
    MCBSP_FSETS(RCR1, RWDLEN1, 32BIT);
    MCBSP_FSETS(XCR1, XWDLEN1, 32BIT);

    DSK6713_AIC23_setFreq(hCodec, DSK6713_AIC23_FREQ_48KHZ);    // set the sampling rate

    // Interrupt setup
    IRQ_globalDisable();            // Globally disables interrupts
    IRQ_nmiEnable();                // Enables the NMI interrupt
    IRQ_map(IRQ_EVT_RINT1,15);      // Maps an event to a physical interrupt
    IRQ_enable(IRQ_EVT_RINT1);      // Enables the event
    IRQ_globalEnable();             // Globally enables interrupts

    while(1)
    {
    }
}
```

See source code in project Lab02

Setting the Sampling Rate

Here we open the codec with the default configuration:

```
hCodec = DSK6713_AIC23_openCodec(0, &config);          // Open the codec
```

The structure “config” is declared in **dsk6713_aic23.h**

Rather than editing the header file, we can change the sampling frequency after the initial configuration:

```
DSK6713_AIC23_setFreq(hCodec, DSK6713_AIC23_FREQ_48KHZ);    // set the sampling rate
```

Frequency definitions are in **dsk6713_aic.h**

```
/* Frequency Definitions */
#define DSK6713_AIC23_FREQ_8KHZ        1
#define DSK6713_AIC23_FREQ_16KHZ       2
#define DSK6713_AIC23_FREQ_24KHZ       3
#define DSK6713_AIC23_FREQ_32KHZ       4
#define DSK6713_AIC23_FREQ_44KHZ       5
#define DSK6713_AIC23_FREQ_48KHZ       6
#define DSK6713_AIC23_FREQ_96KHZ       7
```

Other Codec Configuration

- Input volume (individually controllable for left and right channels)
- Headphone output volume (individually controllable for left and right channels)
- Digital word size (16, 20, 24, or 32 bit)
- Other settings, e.g. byte order, etc. For more details, see:
  - **dsk6713_aic23.h**
  - Codec datasheet (TLV320AIC23B)
  - C:\CCStudio_v3.1\docs\hlp\c6713dsk.hlp

Some Things to Try

- Make a new project that:
  - Polls DIP switch 0. If pressed, light up all four LEDs.
  - Sets the sampling rate of the AIC23 codec to 44.1kHz.
  - Uses an ISR to sample the left and right channels.
  - Multiplies the left and right channels by a variable gain.
  - Outputs the modified samples to the left and right channels.
- Bonus: Swap the channels, i.e. Left_in -> Right_out, Right_in -> Left_out, when DIP switch 0 is pressed.
- Bonus: Try changing the input/output volumes (hint: look at default configuration in dsk6713_aic23.h)

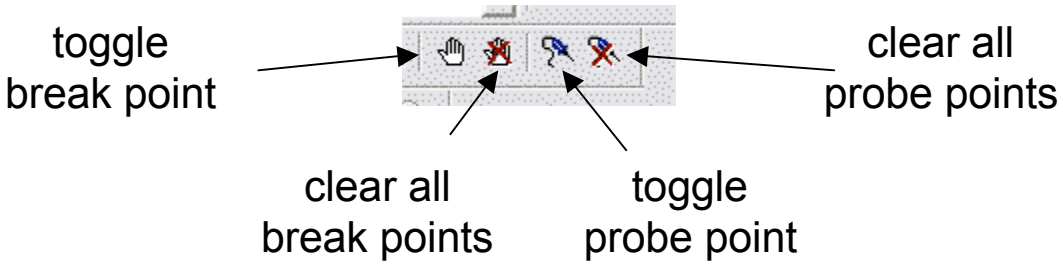
Lunch Break

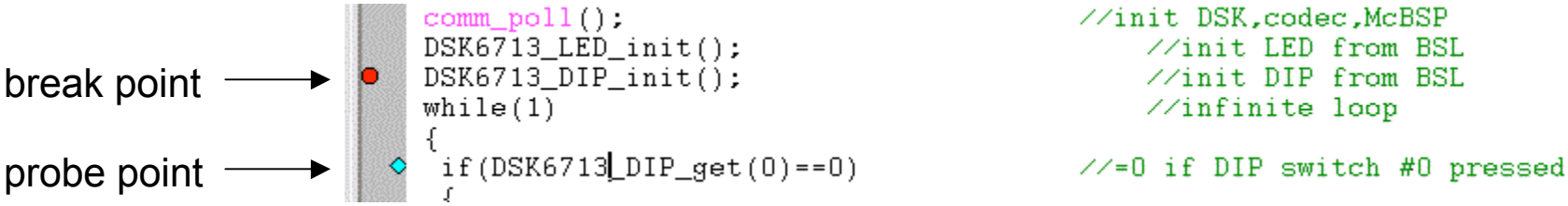
Workshop resumes at 1:30pm…

Debugging and Other Useful Features of the CCS IDE

- Breakpoints
- Probe points
- Watch variables
- Plotting arrays of data
- Animation
- General Extension Language (GEL)

Breakpoints and Probe Points



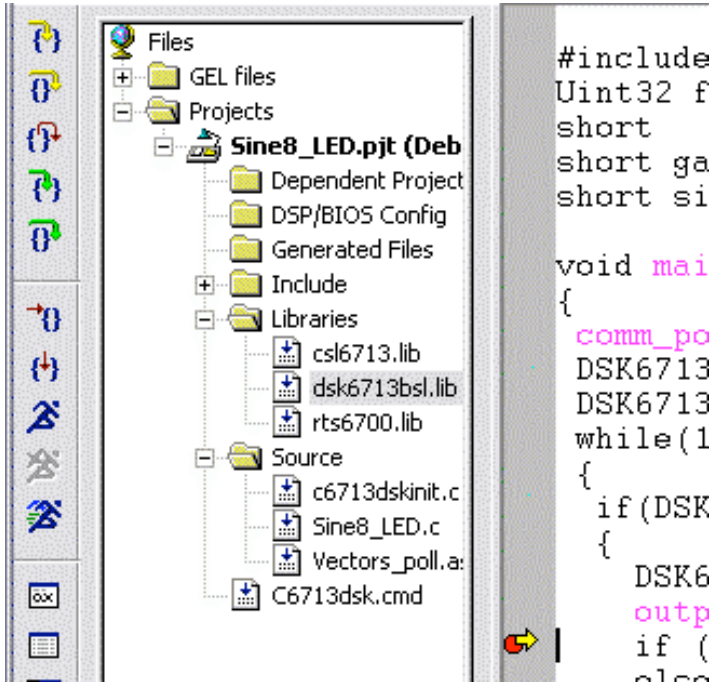


- **Breakpoints**: stop code execution at this point to allow state examination and step-by-step execution.
- **Probe points**: force window updates and/or read/write samples from/to a file at a specific point in your code.

Breakpoints

source step into
source step over
step out
ASM step into
ASM step over

run to cursor
set progam counter to cursor

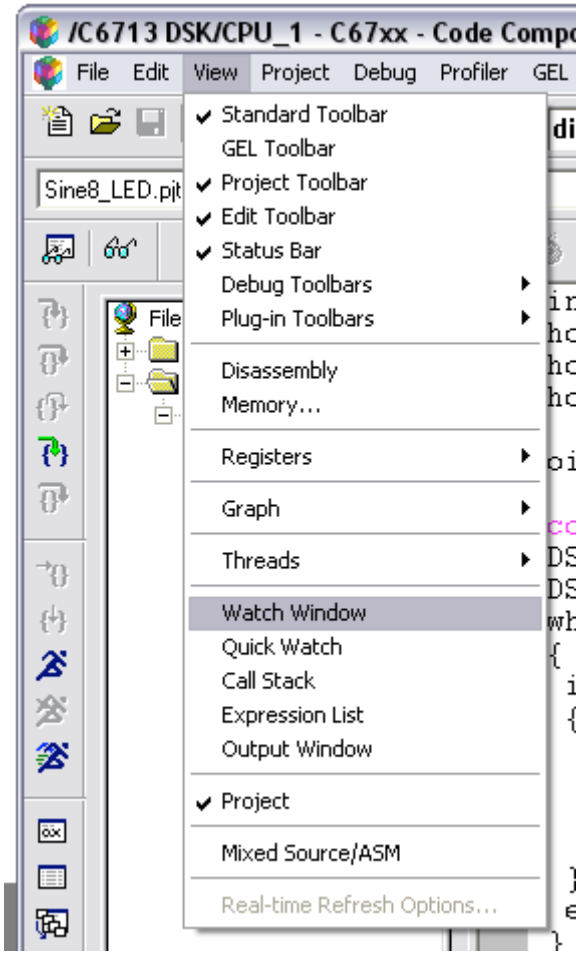


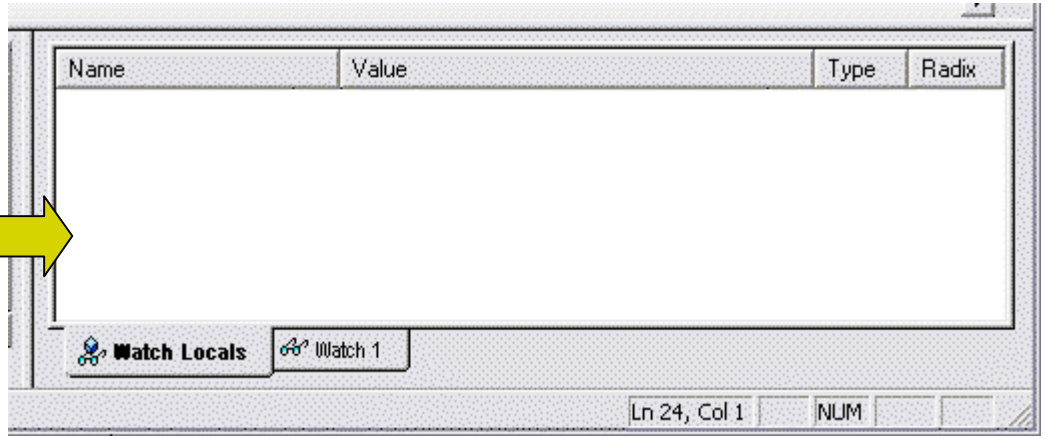
"Run to Cursor" is a handy shortcut instead of setting a breakpoint

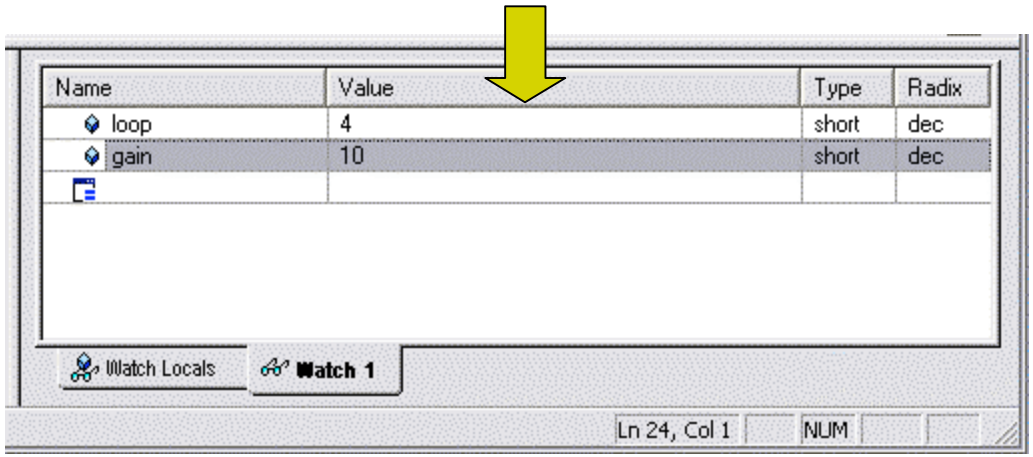
Probe Points

- Differ from breakpoints: Halt the DSP momentarily, perform an action, and then automatically resume execution.
  - Note that this may cause problems with real-time operations.
- Facilitate repeatable testing via automatic file input and/or output (on PC).
- For more details, see CCS Getting Started Guide (SPRU509F.PDF) or CCS help.

Watch Variables



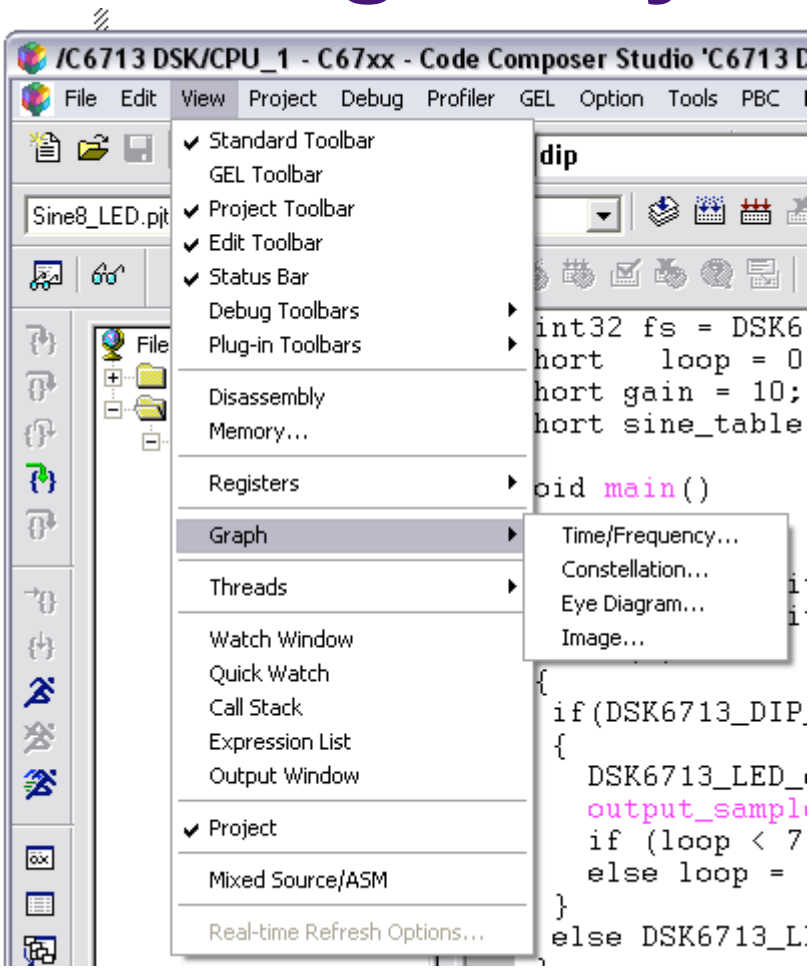


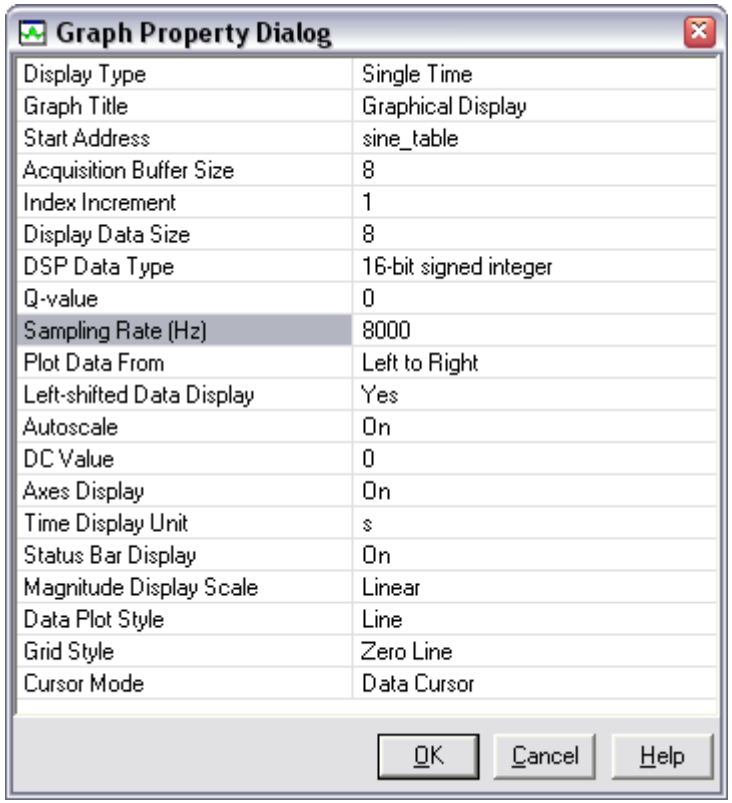


Watch Variables

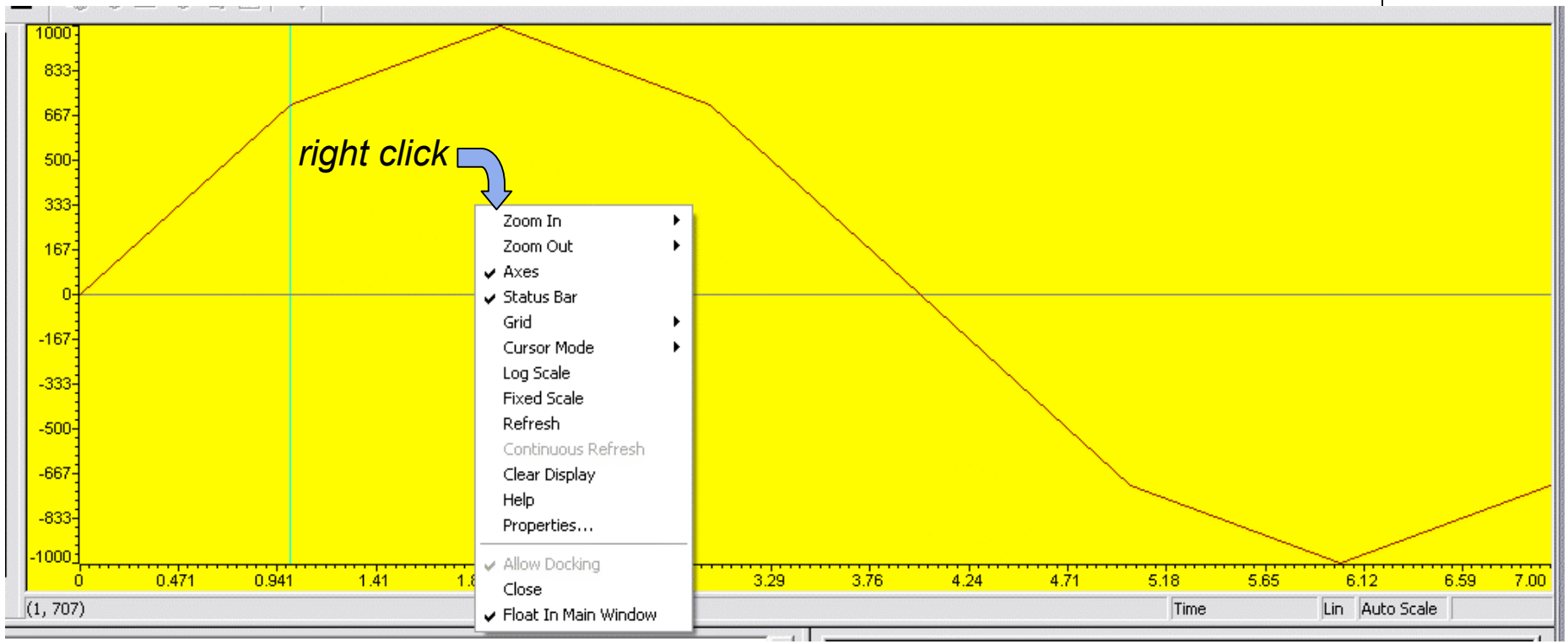
- In the **Watch Locals** tab, the debugger *automatically* displays the Name, Value, and Type of the variables that are *local* to the currently executing function.
- In the **Watch** tab, the debugger displays the Name, Value, and Type of the local and global variables and expressions that *you specify*.
- Can add/delete tabs.

Plotting Arrays of Data





Plotting Arrays of Data

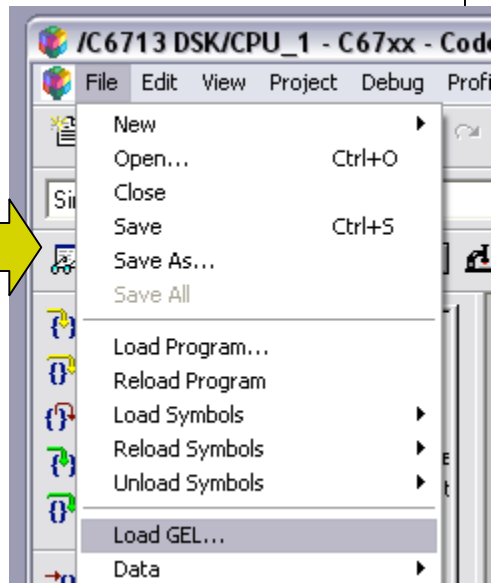


Animation

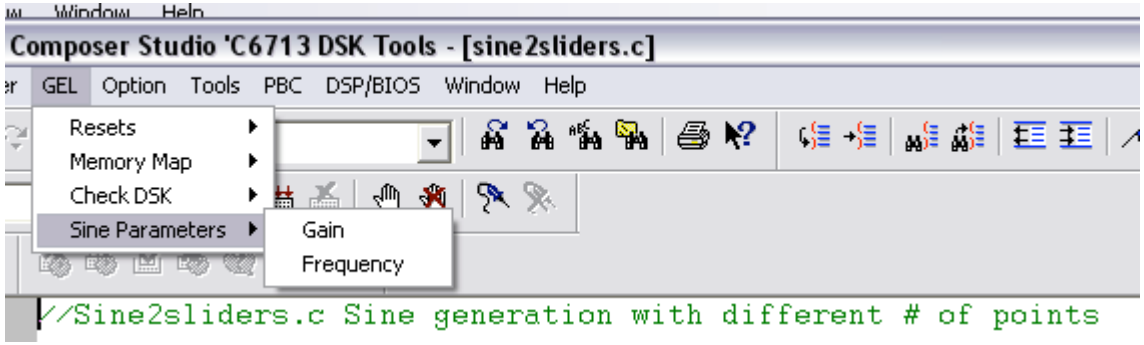
- Runs the program until a breakpoint is encountered.
  - At the breakpoint, execution stops and all windows not connected to any Probe Points are updated.
  - Program execution then automatically resumes
  - Useful for updating graphical displays
  - Note: Animation may cause problems with real-time operation
- Can pause execution at each breakpoint: **Option->Customize: Debug Properties** tab Animate Speed (0-9s) (zero = no pause)

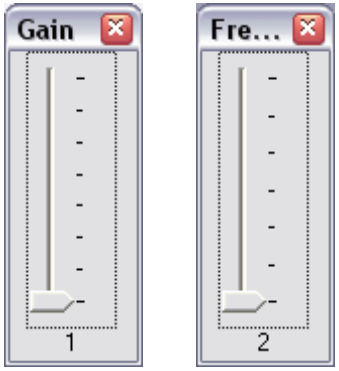
General Extension Language

- Create functions to extend the functionality of Code Composer Studio
- GEL files are not loaded with a project



- Often used to change variables "on-the-fly"
- Examples from Chassaing textbook: **sin2sliders.pjt** and **sin2sliders.gel**





General Extension Language

- Useful GEL files can be pretty simple
- From **sin2sliders.gel:**

```
/*Sine2sliders.gel Two sliders to vary gain and frequency*/

menuitem "Sine Parameters"

slider Gain(1,8,1,1,gain_parameter)          /*incr by 1,up to 8*/
    {
        gain = gain_parameter;               /*vary gain*/
    }

slider Frequency(2,8,2,2,frequency_parameter) /*incr by 2,up to 8*/
    {
        frequency = frequency_parameter;      /*vary frequency*/
    }
```

- Syntax details can be found in CCS help: **Help->Contents->Making a Code Composer Studio Project -> Building and Running your Project -> Automating Tasks with General Extension Language (GEL)**

Some Things to Try

- Try out the debugging tools on the code you wrote in the morning session
  - breakpoints
  - probe points
  - watch variables
  - animation
- Modify your stereo in/out project to have the output gain changeable via a GEL slider
- Try out the CCS plotting tools
  - Modify your code to have a buffer (i.e., store samples in an array) and plot the contents.
- Try to have CCS animate a plot window

Finite Impulse Response (FIR) Filters

- Frequently used in real-time DSP systems
  - Simple to implement
  - Guaranteed to be stable
  - Can have nice properties, e.g. linear phase
- Input/output relationship

$$y[n] = \sum_{m=0}^{M-1} h[m]x[n-m]$$

x=input, ***y***=output, ***h***=filter coefficients, ***M***=# of filter coefficients

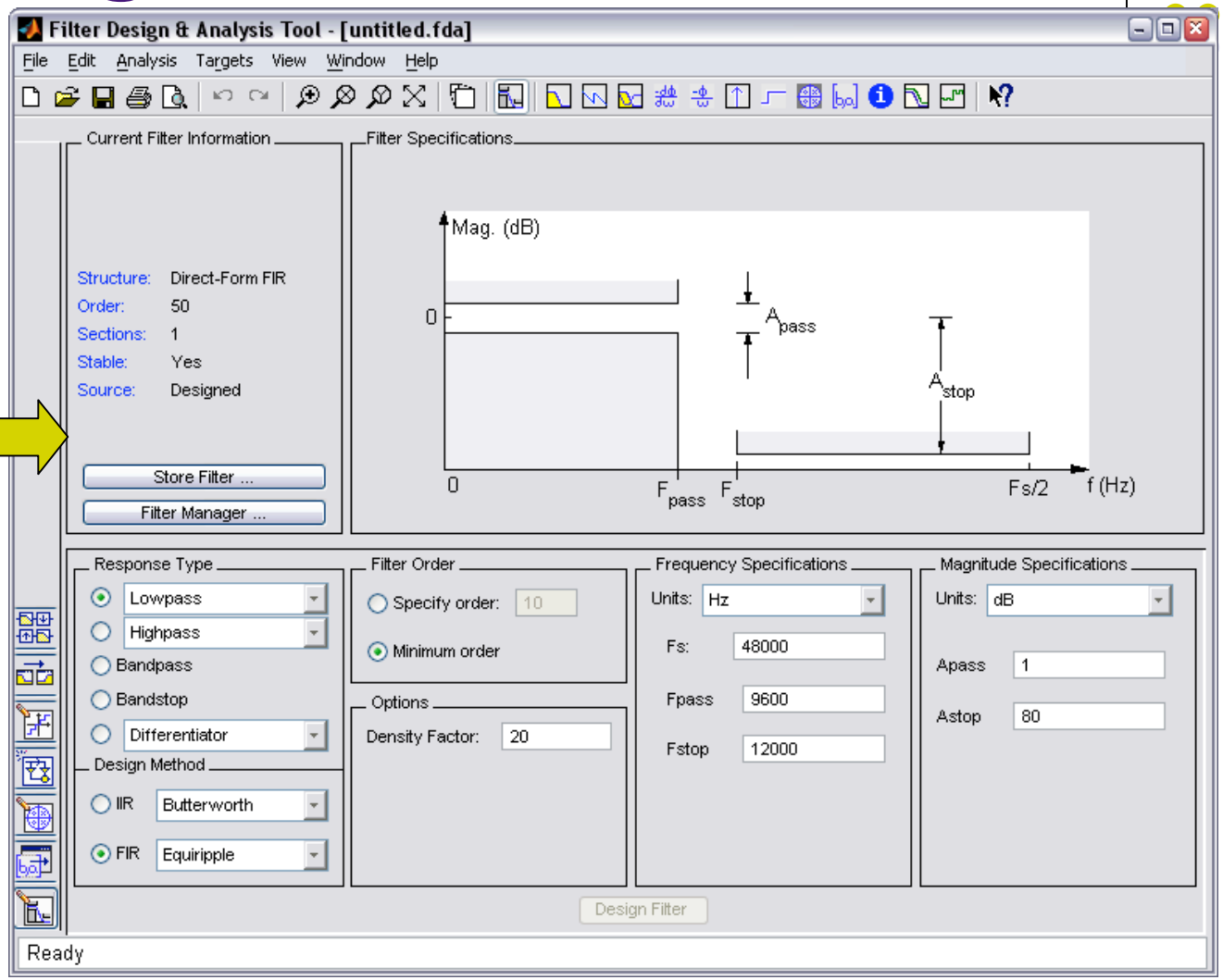
Creating FIR Filters

1. Design filter — ***Matlab***
   - Type: low pass, high pass, band pass, band stop, ...
   - Filter order M
   - Desired frequency response
2. Decide on a realization structure
3. Decide how coefficients will be quantized.
4. Compute quantized coefficients

5. Decide how everything else will be quantized (input samples, output samples, result of multiplies, result of additions) — ***CCS***
6. Write code to realize filter
7. Test filter and compare to theoretical expectations

Designing FIR Filters

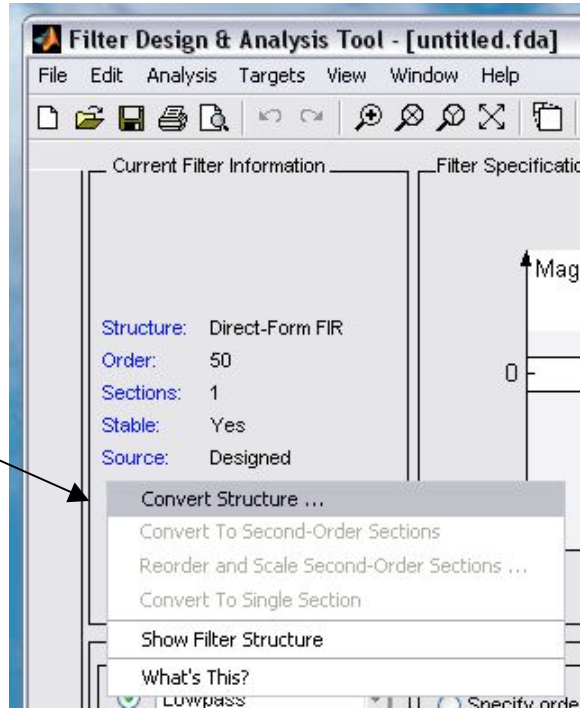
>> fdatool

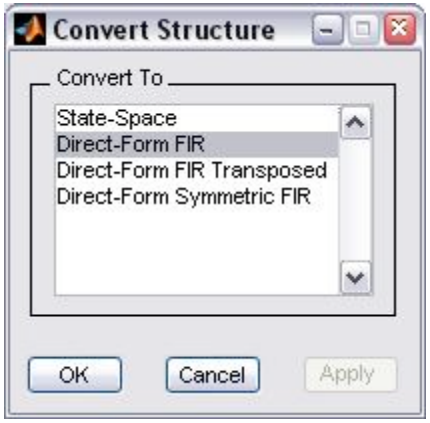


Filter Realization Structures

- Lots of different structures available
  - Direct form I, direct form II, transposed forms, cascade, parallel, lattice, …
  - All have same input/output relationship
  - Choice of structure affects computational complexity and how quantization errors are manifested through the filter

right click in this pane





Focus on "Direct form" for now. We'll discuss other options when we look at IIR filtering tomorrow.

Compute FIR Filter Coefficients

Filter Design & Analysis Tool - [untitled.fda]

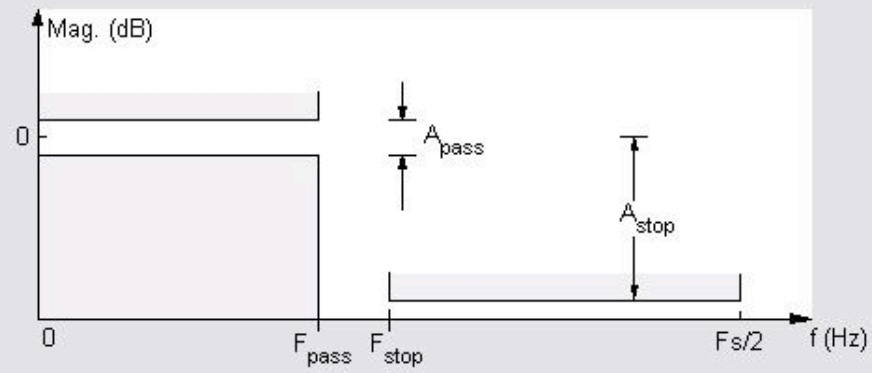
File Edit Analysis Targets View Window Help

Current Filter Information

Structure: Direct-Form FIR
Order: 50
Sections: 1
Stable: Yes
Source: Designed

Store Filter ...
Filter Manager ...

Filter Specifications



Response Type
- Lowpass
- Highpass
- Bandpass
- Bandstop
- Differentiator

Design Method
- IIR Butterworth
- FIR Equiripple

Filter Order
- Specify order: 10
- Minimum order

Options
- Density Factor: 20

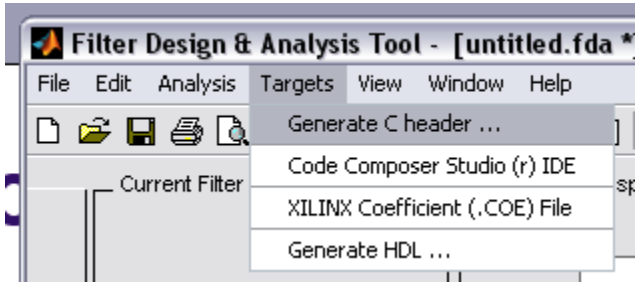
Frequency Specifications
- Units: Hz
- Fs: 48000
- Fpass 9600
- Fstop 12000

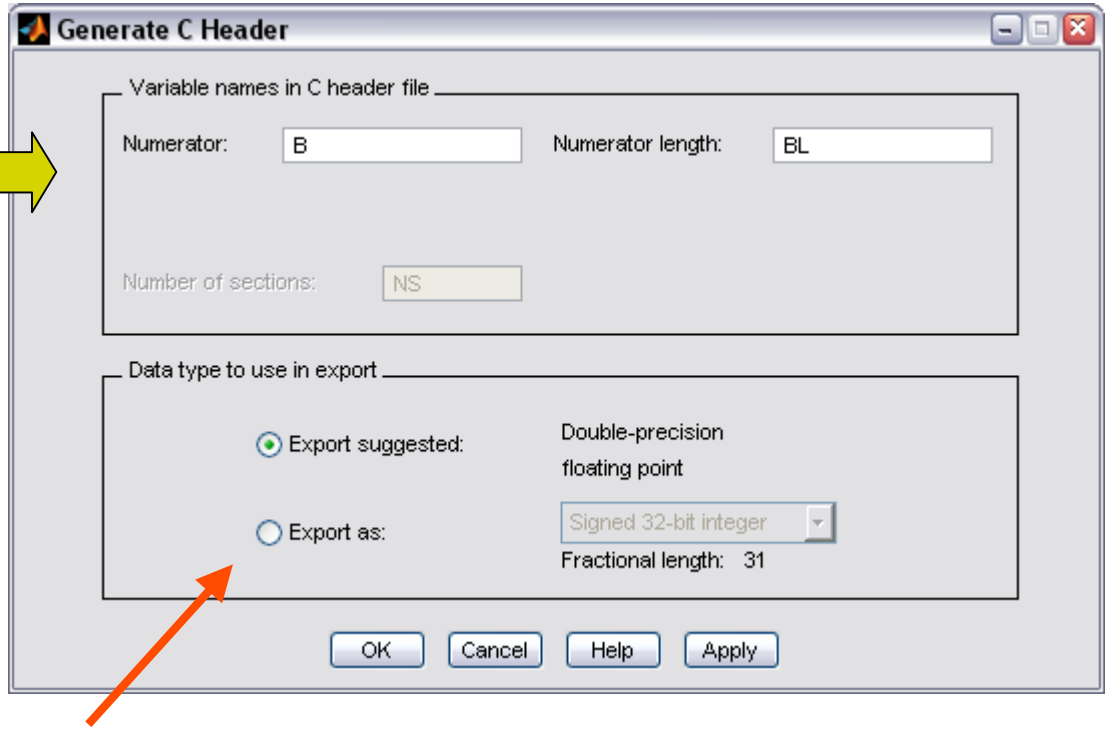
Magnitude Specifications
- Units: dB
- Apass 1
- Astop 80

set up filter and press → Design Filter

Ready

Make Coefficient File For CCS





Here you can change the coefficient data type to match your desired quantization.

Example DP-FP Coefficient File

```
/*
 * Filter Coefficients (C Source) generated by the Filter Design and Analysis Tool
 *
 * Generated by MATLAB(R) 7.0 and the
 *
 * Generated on: 19-Aug-2005 13:04:09
 *
 */

/*
 * Discrete-Time FIR Filter (real)
 * -------------------------------
 * Filter Structure   : Direct-Form FIR
 * Filter Order       : 8
 * Stable             : Yes
 * Linear Phase       : Yes (Type 1)
 */

/* General type conversion for MATLAB generated C-code  */
#include "tmwtypes.h"
/*
 * Expected path to tmwtypes.h
 * C:\MATLAB7\extern\include\tmwtypes.h
 */
const int BL = 9;
const real64_T B[9] = {
    0.02588139692752,  0.08678803067191,   0.1518399865268,   0.2017873498839,
     0.2205226777929,   0.2017873498839,   0.1518399865268,  0.08678803067191,
    0.02588139692752
};
```

Can edit these to agree with your code.

Quantization Considerations

- Key choice: **floating point** vs. **fixed point**
- Advantages of floating point math:
  - Less quantization error
  - Don't have to worry about scaling factors
  - Less likelihood of overflow/underflow
  - Much easier to code
- Disadvantages of floating point math:
  - Requires floating point DSP (higher cost, higher power)
  - Executes slower than fixed point
- C code allows you to "cast" variables into any datatype

Write Code to Realize FIR Filter

- Direct form I implies direct realization of the convolution equation

$$y[n] = \sum_{m=0}^{M-1} h[m]x[n-m]$$

- Some considerations:
  - Allocate buffer of length M for input samples.
  - Move input buffer pointer as new data comes in or move data?

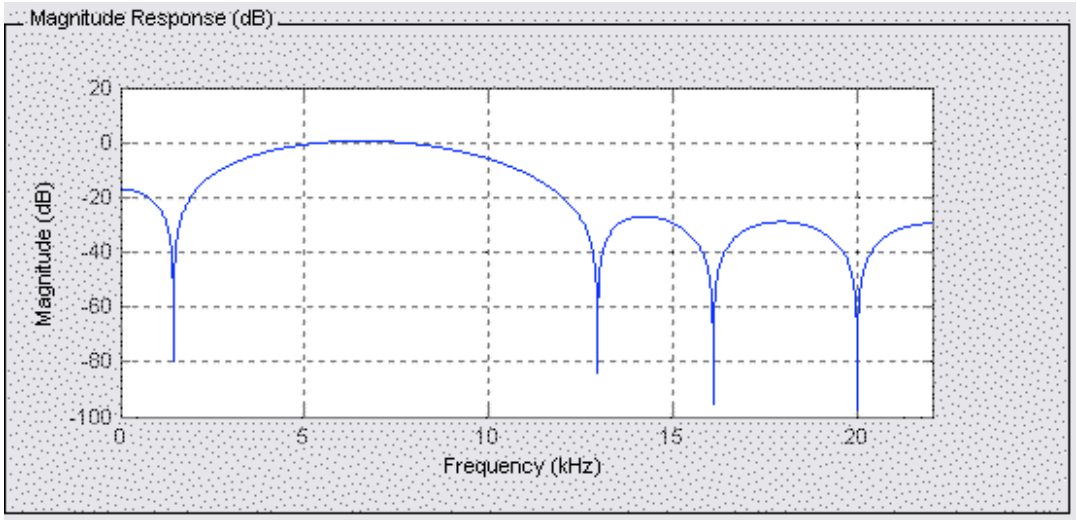
FIR filter example Code

```
interrupt void serialPortRcvISR()
{
        union {Uint32 combo; short channel[2];} temp;
        int i = 0;
        float result = 0.0;

        temp.combo = MCBSP_read(DSK6713_AIC23_DATAHANDLE);

        // Update array samples (move data - this is the slow way)
        for( i = N-1; i >= 1; i-- )
                samples[i] = samples[i-1];
        samples[0] = (float)temp.channel[0];   // store right channel

        // Filtering
        for( i = 0 ; i < N ; i++ )
                result += fir_coeff[i]*samples[i];
        temp.channel[0] = (short)result; // output to right channel
        MCBSP_write(DSK6713_AIC23_DATAHANDLE, temp.combo);
}
```

Note that all math here is floating point.
Filter coefficients are also assumed to be floating point.

Some Things to Try

- Try creating an FIR filter with the following specs:
  - Bandpass
  - 8th order Direct Form I
  - Least-squares design
  - 44100Hz sampling rate
  - Fstop1 = 3000Hz
  - Fpass1 = 4000Hz
  - Fpass2 = 8000Hz
  - Fstop2 = 12000Hz
  - Equal weighting in all bands
  - All floating point math (single or double precision)



- Use an oscilloscope and a function generator to compare the magnitude response of your filter to the theoretical prediction.

Workshop Day 1 Summary

What you learned today:

- Basics of the TMS320C6713 DSK and Code Composer Studio
- How to test the DSK
- How to open, build, load, and run existing projects
- How to create, build, load, and run new projects
- How to interface with DSK I/O (LEDs, DIP switches, and the AIC23 codec)
- How to debug code in CCS including
  - Setting and clearing breakpoints and probe points
  - Setting up watch variables
  - Plotting arrays of data
  - Animation
- How to use, modify, and create GEL files in CCS.
- How to use Matlab's filter design/analysis tool "fdatool"
- How to implement an FIR filter on the C6713

Workshop Day 1 Reference Material

- Chassaing textbook Chapters 1-2, and 4
- CCS Help system
- **SPRU509F.PDF** CCS v3.1 IDE Getting Started Guide
- **C6713DSK.HLP** C6713 DSK specific help material
- AIC23 Codec datasheet
- DSK Quick Start Guide (included in your DSK box)
- Spectrum Digital TMS320C6713 DSK reference (included in your DSK box)
- TMS320C6000 Programmer's Guide (SPRU198G.PDF)
- Matlab fdatool help (>> doc fdatool)

Latest TI documentation available at http://www.ti.com/sc/docs/psheets/man_dsp.htm