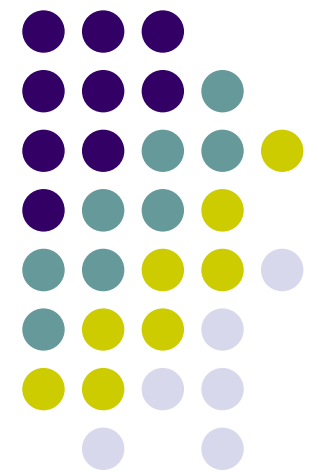


# Digital Signal Processing and Applications with the TMS320C6713 and the TMS3206416 DSK

Day 1

D. Richard Brown III  
Associate Professor  
Worcester Polytechnic Institute  
Electrical and Computer Engineering Department  
drb@ece.wpi.edu

October 16-17, 2006



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
- **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





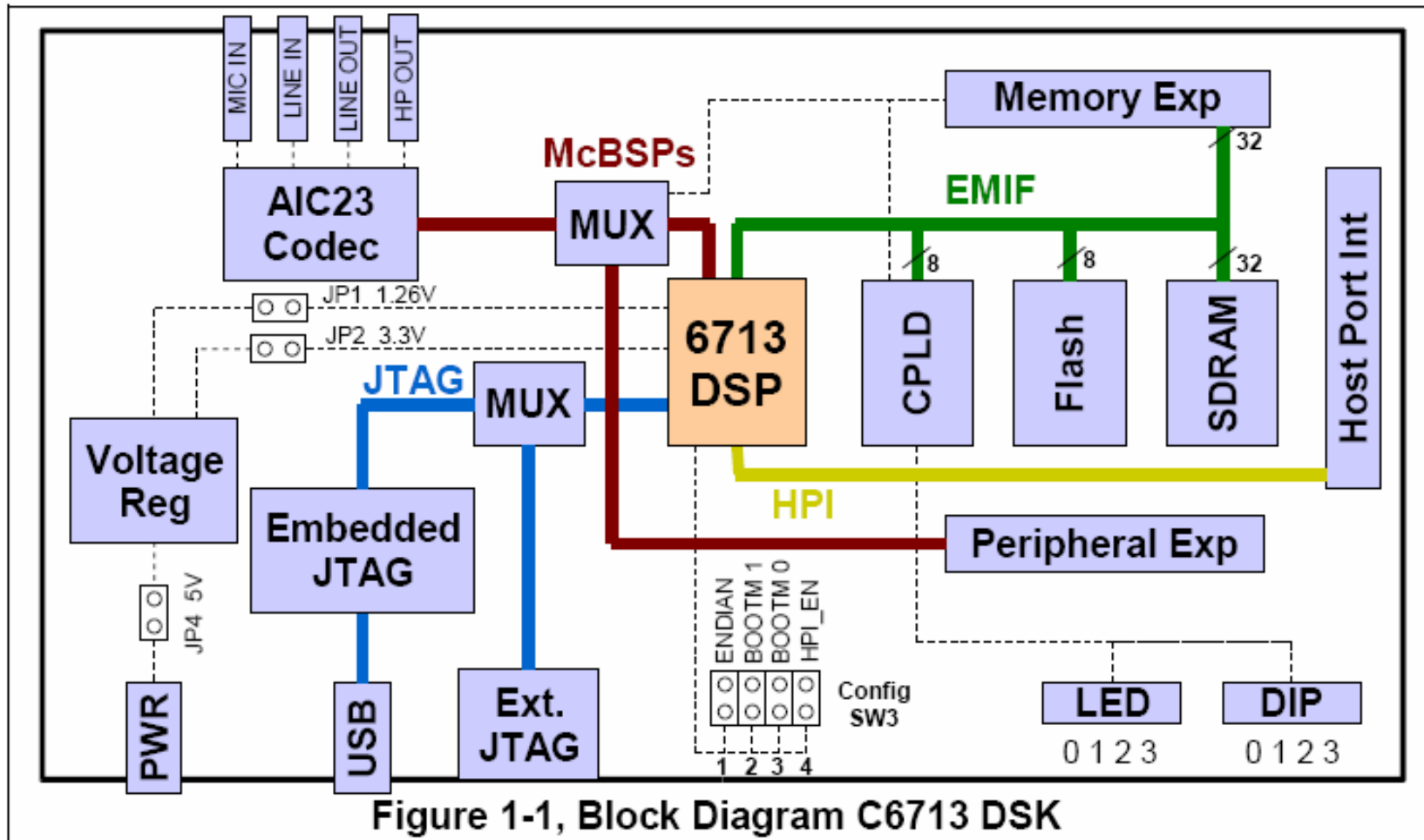
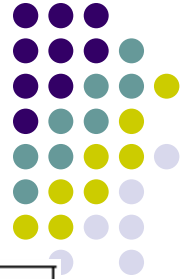
# C6713 DSK Overview

- 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



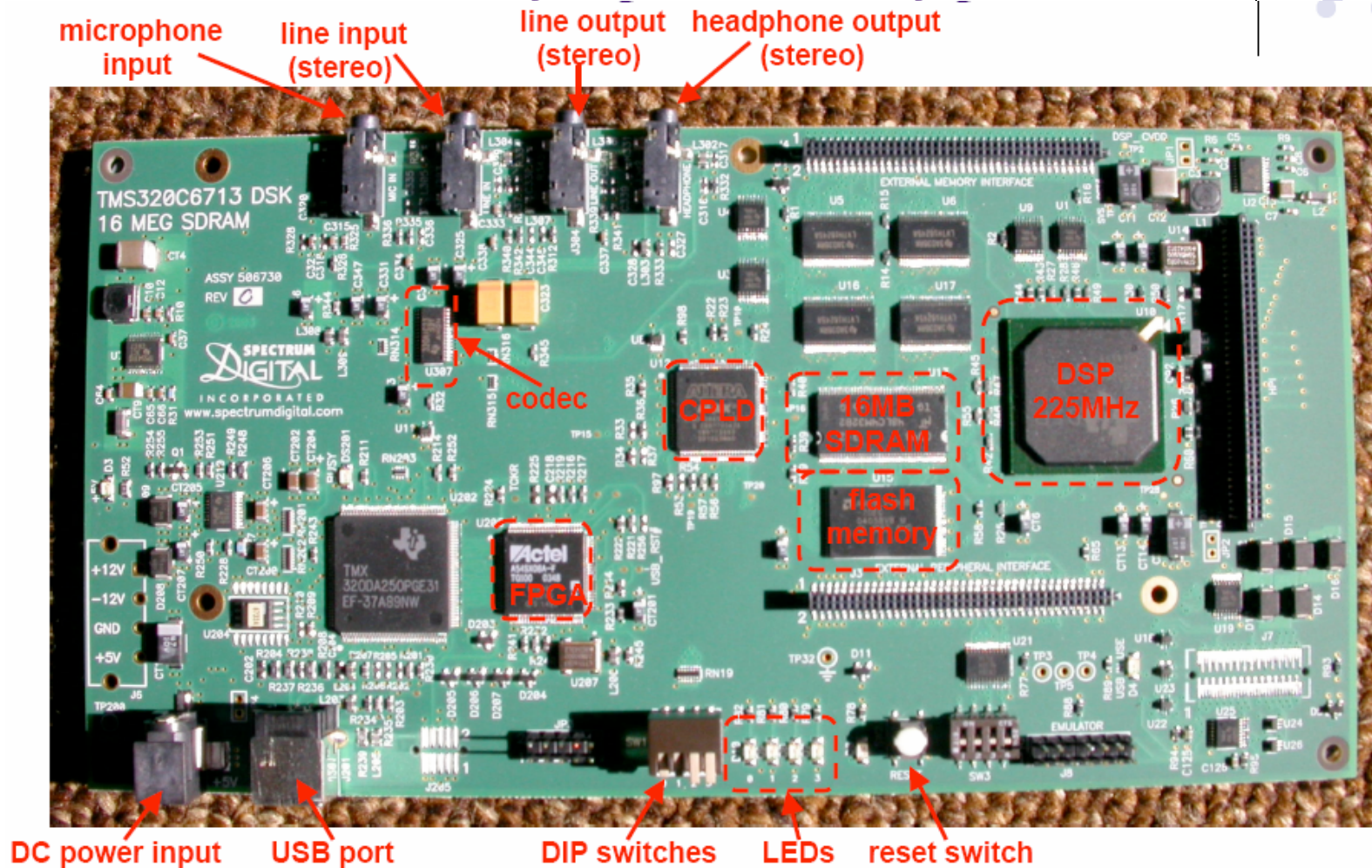
# C6713 DSK

## Functional Block Diagram



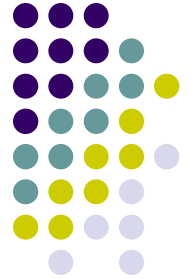


# C6713 DSK Physical Layout



# Is my DSK working?

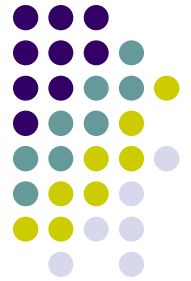
## DSK Power On Self Test



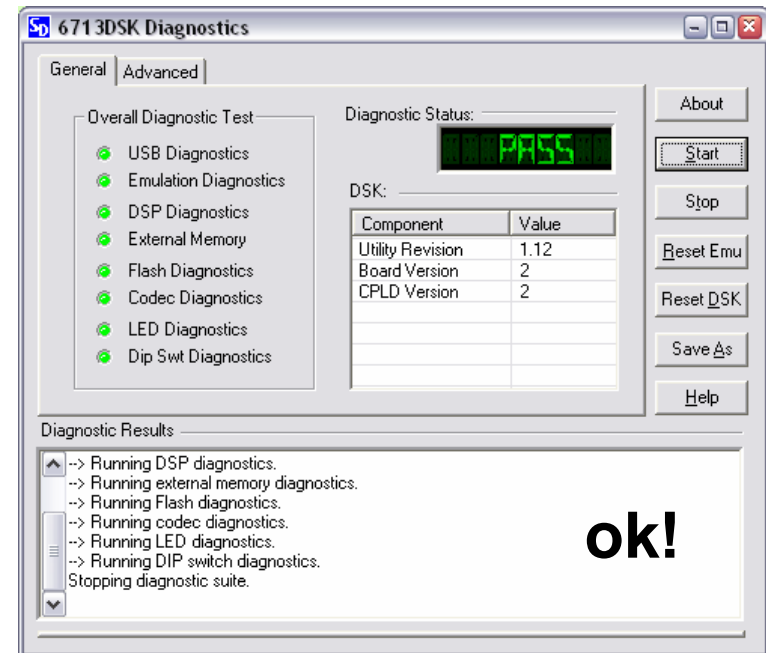
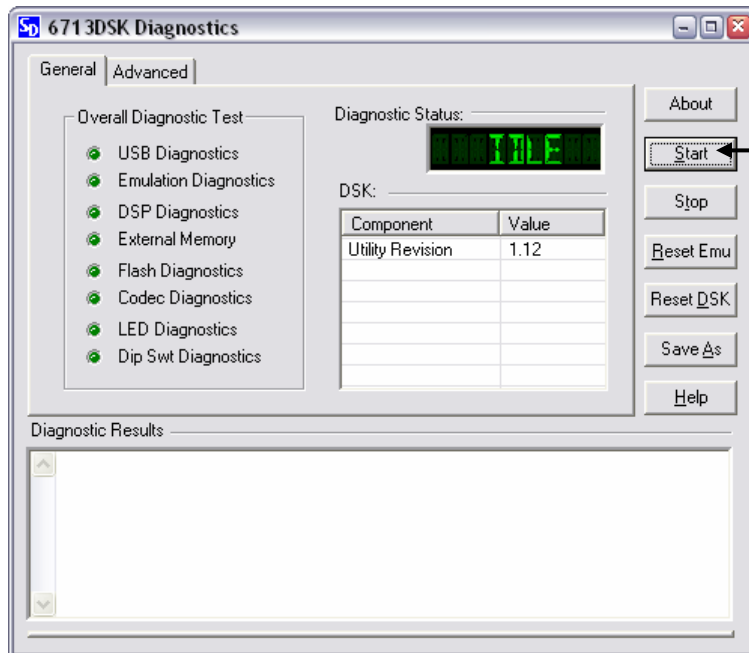
- Power up DSK and watch LEDs
- Power On Self Test (POST) program stored in FLASH memory automatically executes
- POST takes 10-15 seconds to complete
- All DSK subsystems are automatically tested
- During POST, a 1kHz sinusoid is output from the AIC23 codec for 1 second
  - Listen with headphones or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

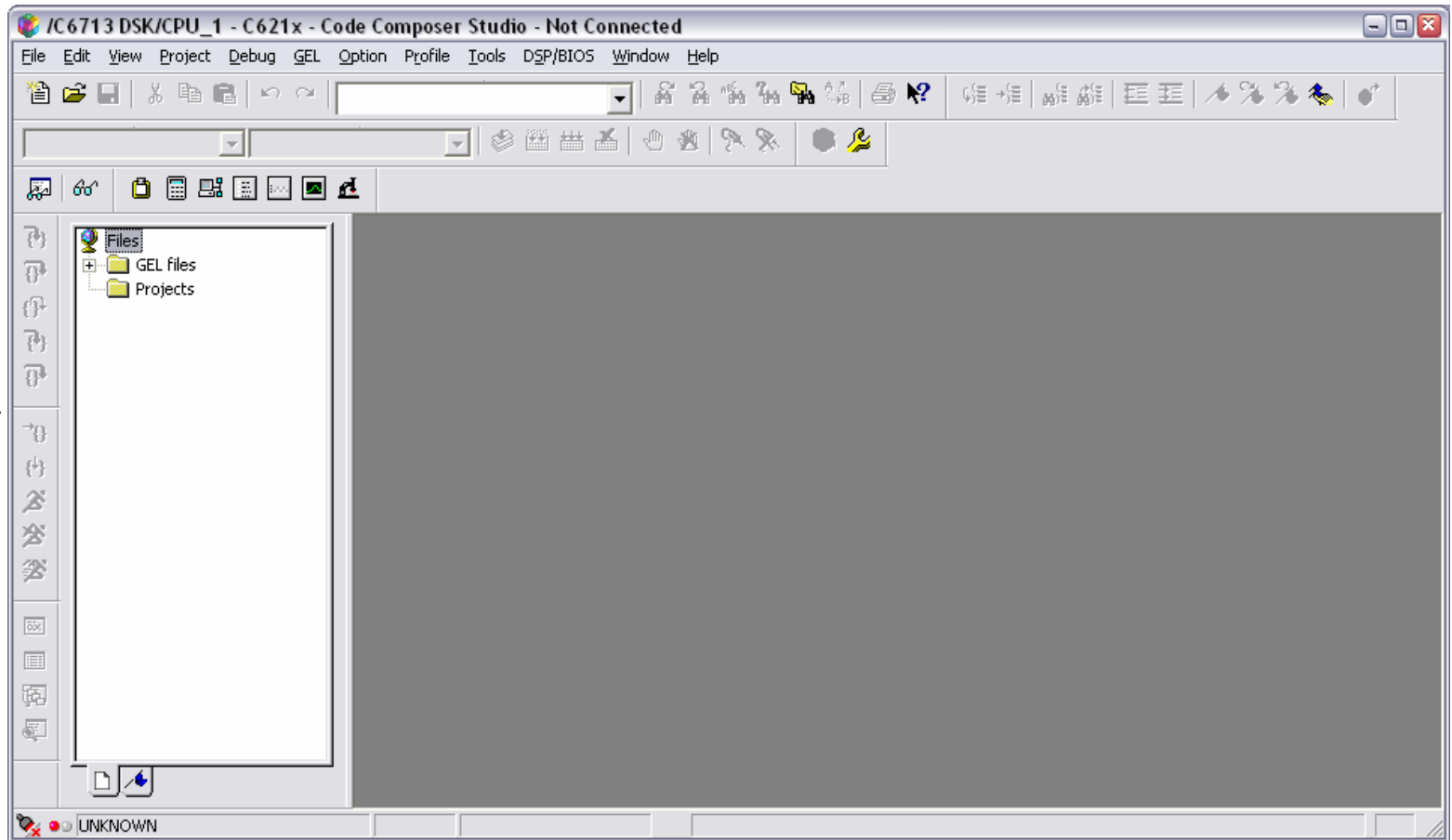
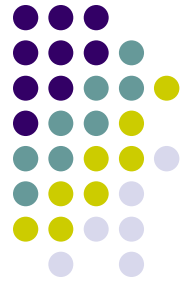




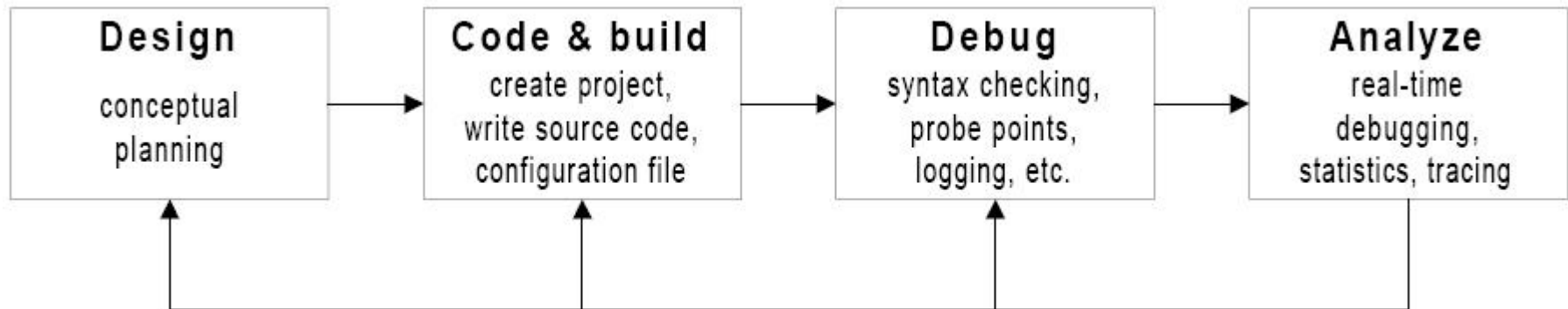
# Code Composer Studio IDE

- 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 (installed to your hard drive):

**SPRU509F.PDF** CCS v3.1 IDE Getting Started Guide

**C6713DSK.HLP** C6713 DSK specific help material

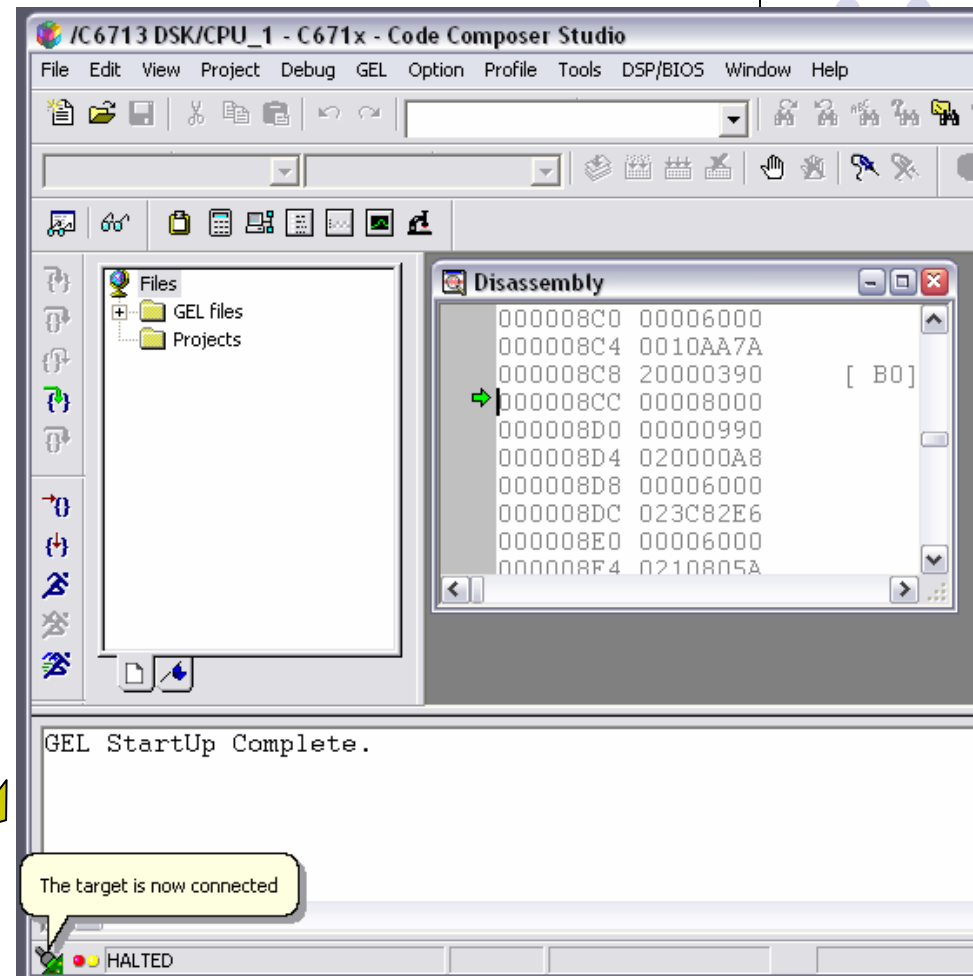
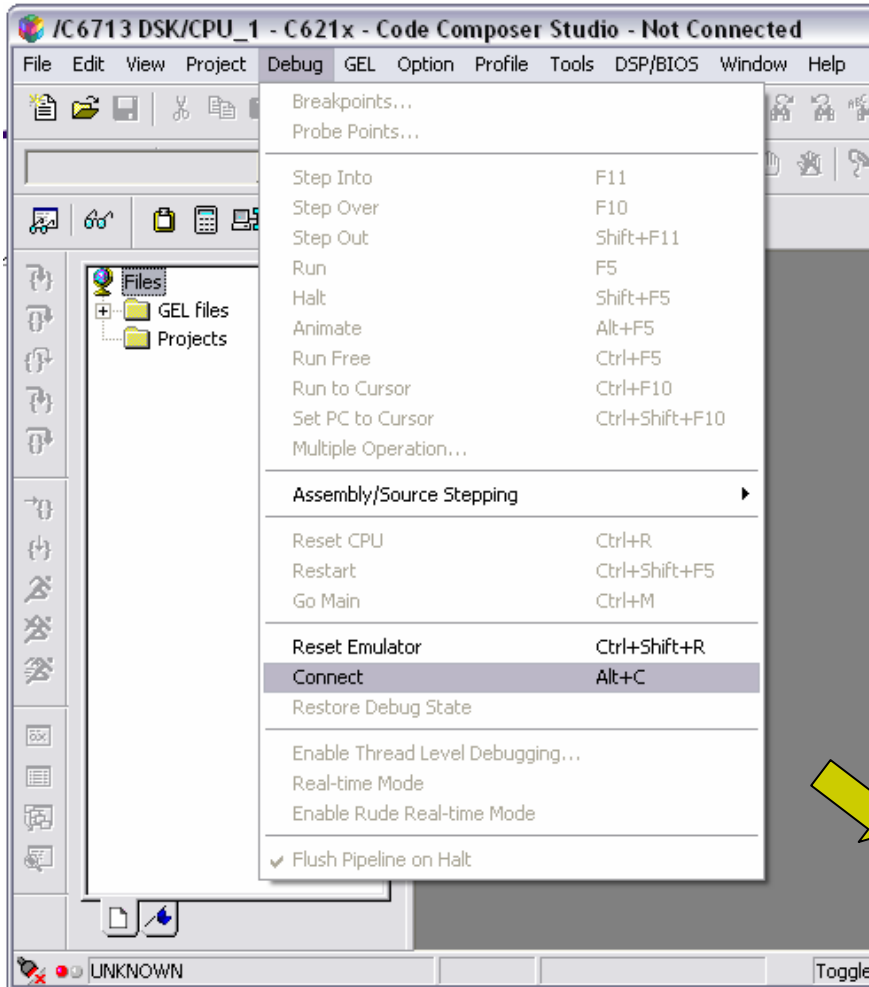
Note that your DSK includes CCS v3.1. Updates and patches are available after registering CCS.



TEXAS INSTRUMENTS

Technology for Innovators™

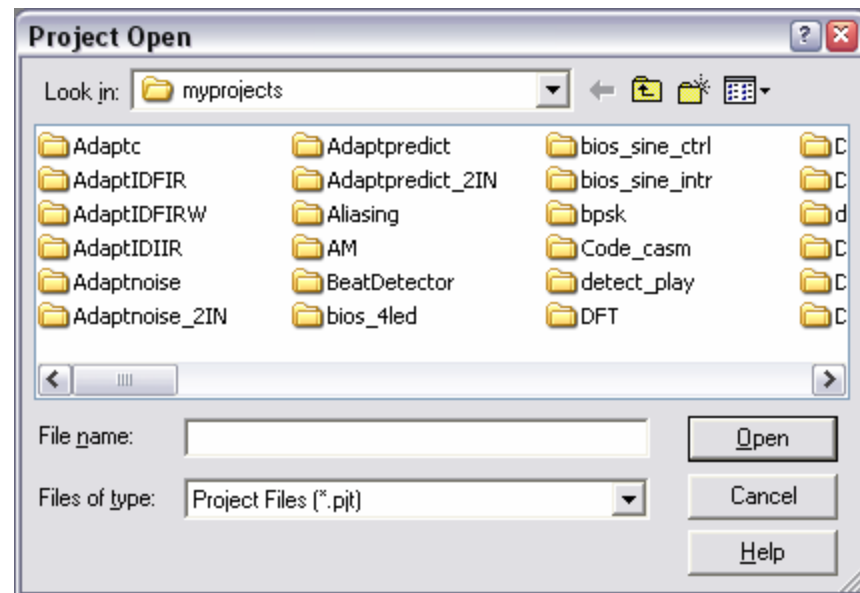
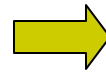
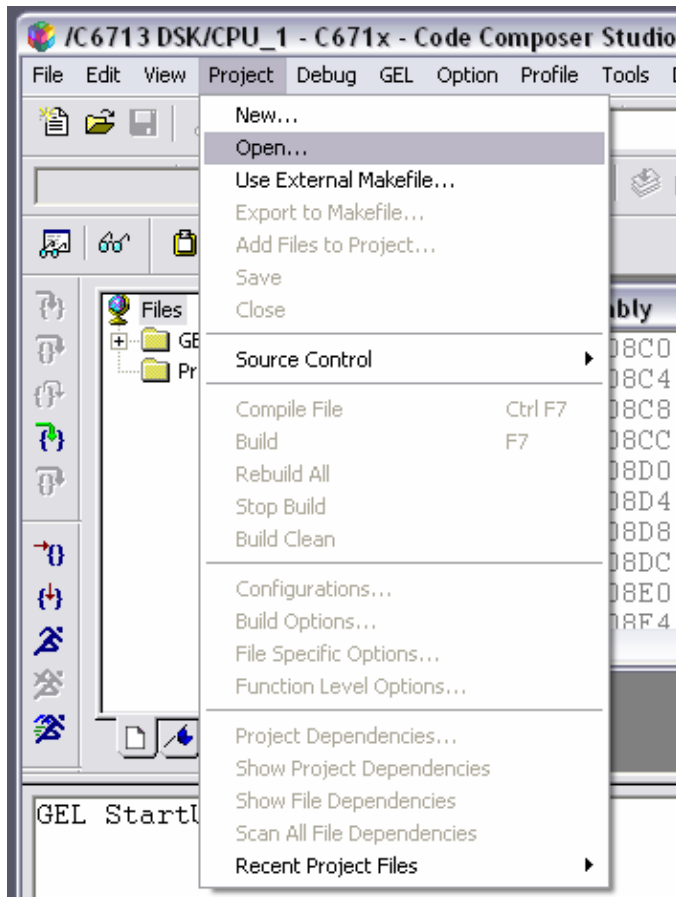
# Connecting to the C6713 DSK





# Opening an Existing Project

Project->Open



Select a .PJT file and press “Open”. Chassaing example projects should be in

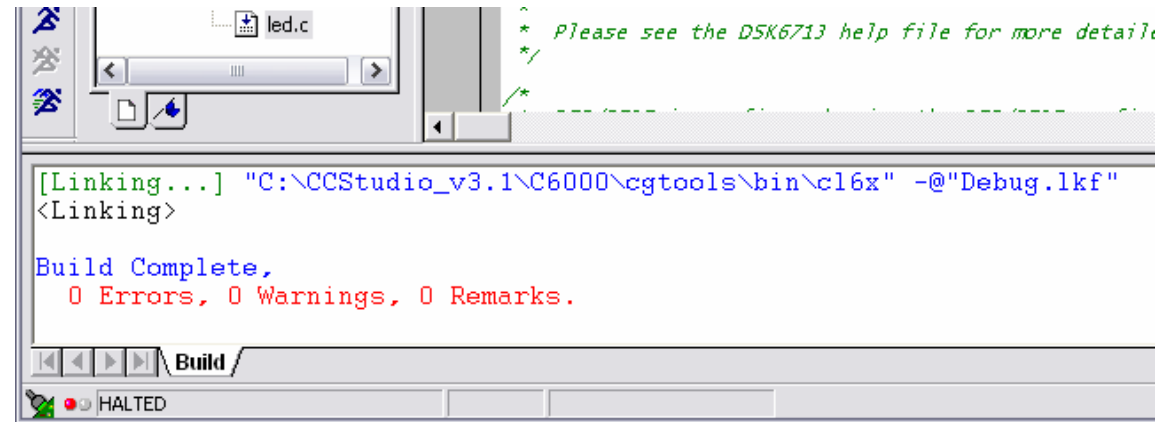
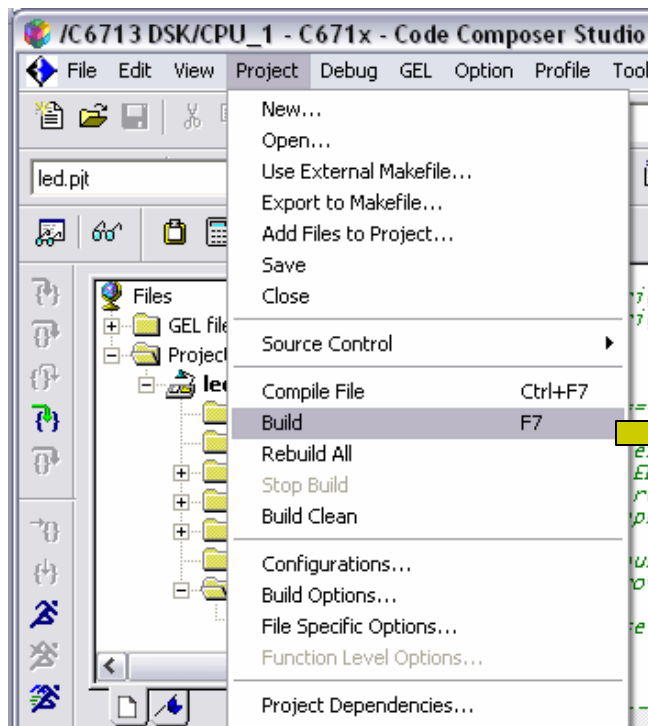
**c:\CCStudio\_v3.1\myprojects\**

Other example projects for the C6713 can be found in **c:\CCStudio\_v3.1\examples\dsk6713**



# Compiling/Building a Project

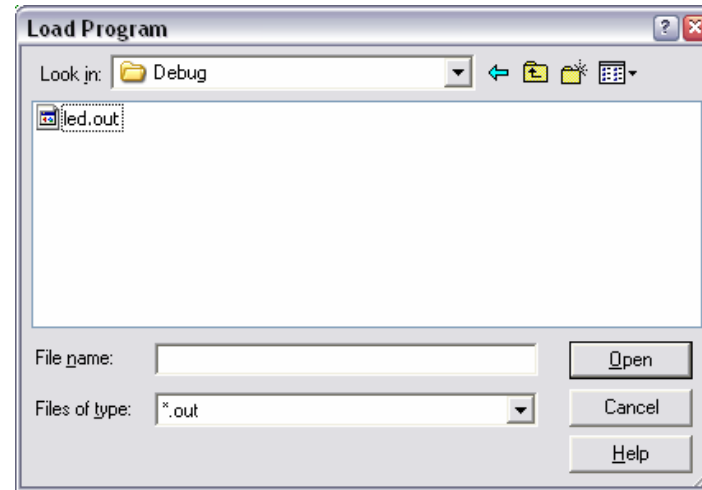
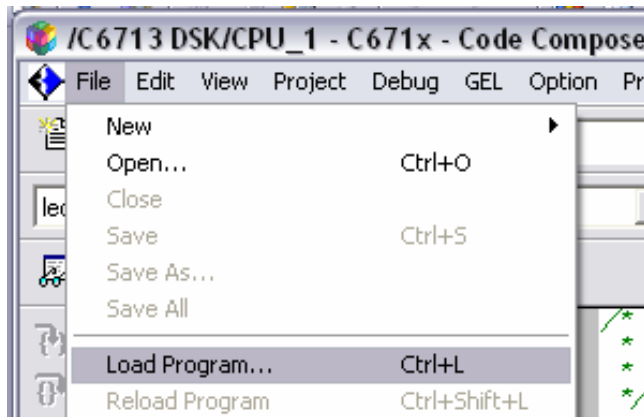
## Project->Build (F7)



# Loading and Running a Project on the C6713 DSK

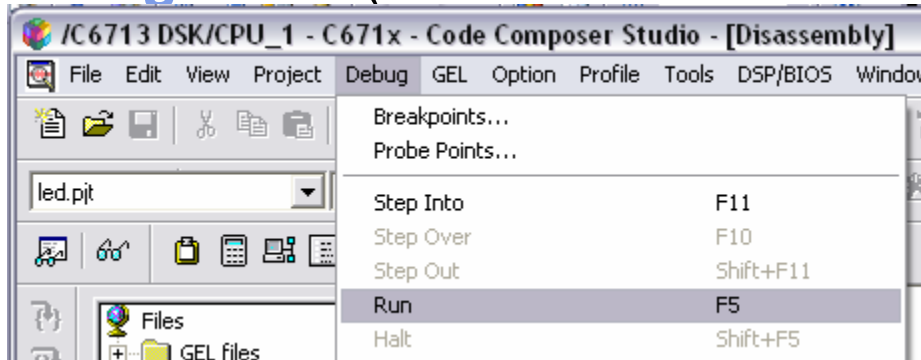


File-> Load Program (ctrl+L)

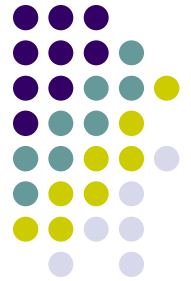



Select the .out file in the project\Debug directory. Program is sent to DSK.

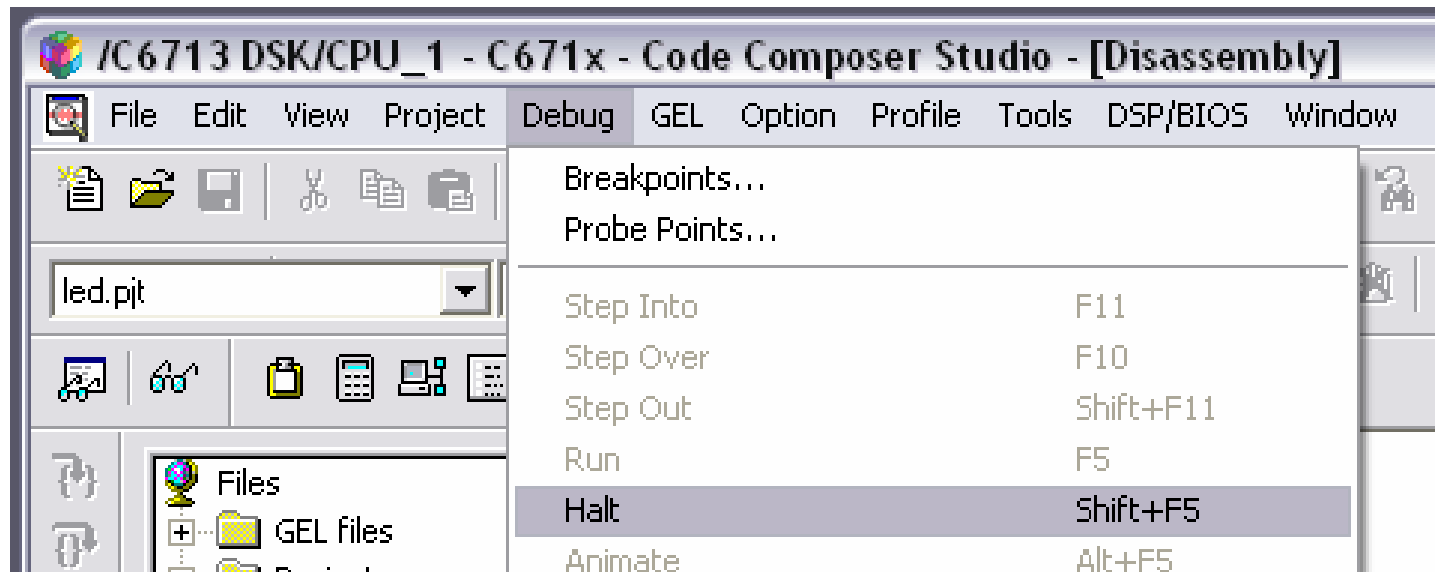
Debug->Run (F5 or the Run button  )



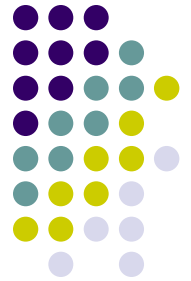
# Halting a Running Program on the C6713 DSK



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

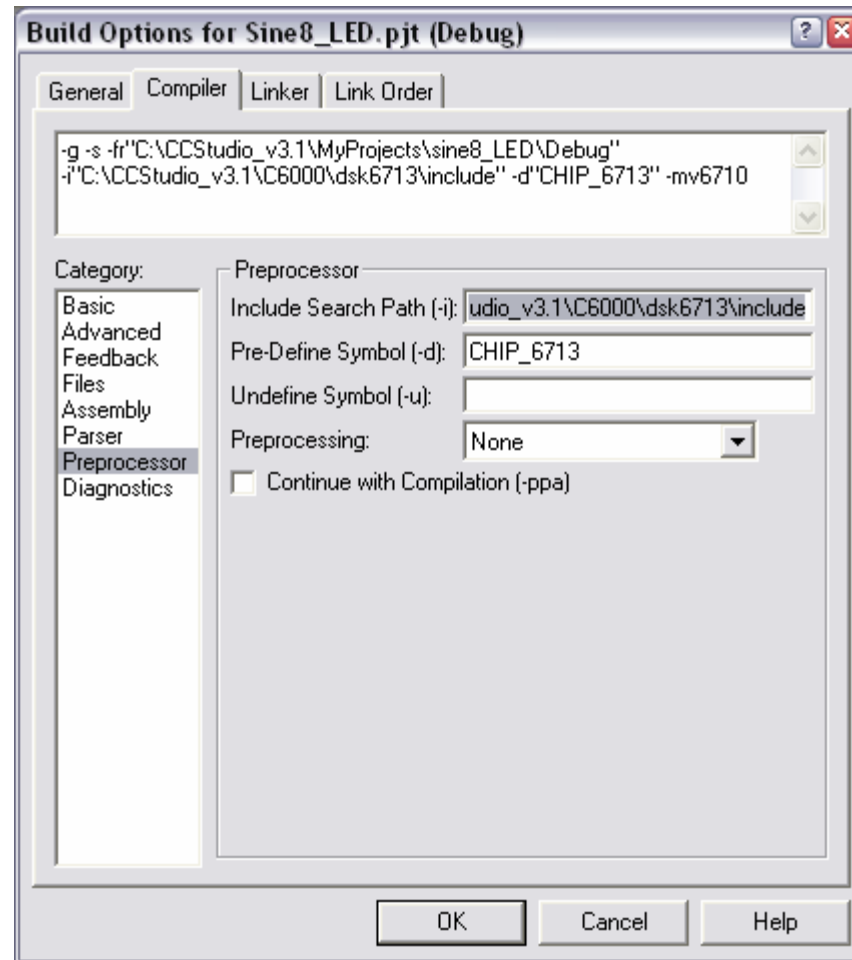


# Chassaing examples: Fixing the search path



Add C:\CCStudio\_v3.1\C6000\dsk6713\include to the search path

Project ->  
Build Options

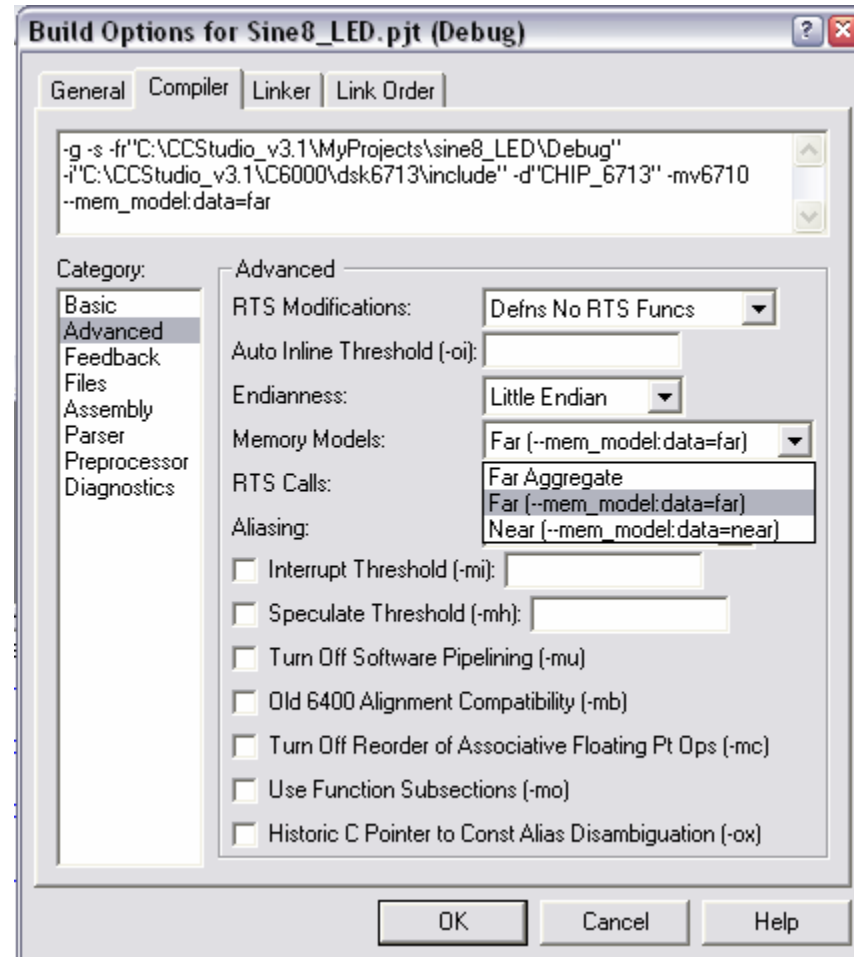


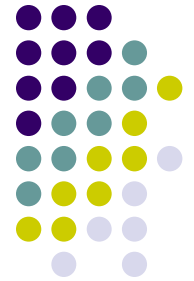


# Chassaing examples: Fixing the mem model

Change the memory model to “data=far”

Project ->  
Build Options





# Things to Try

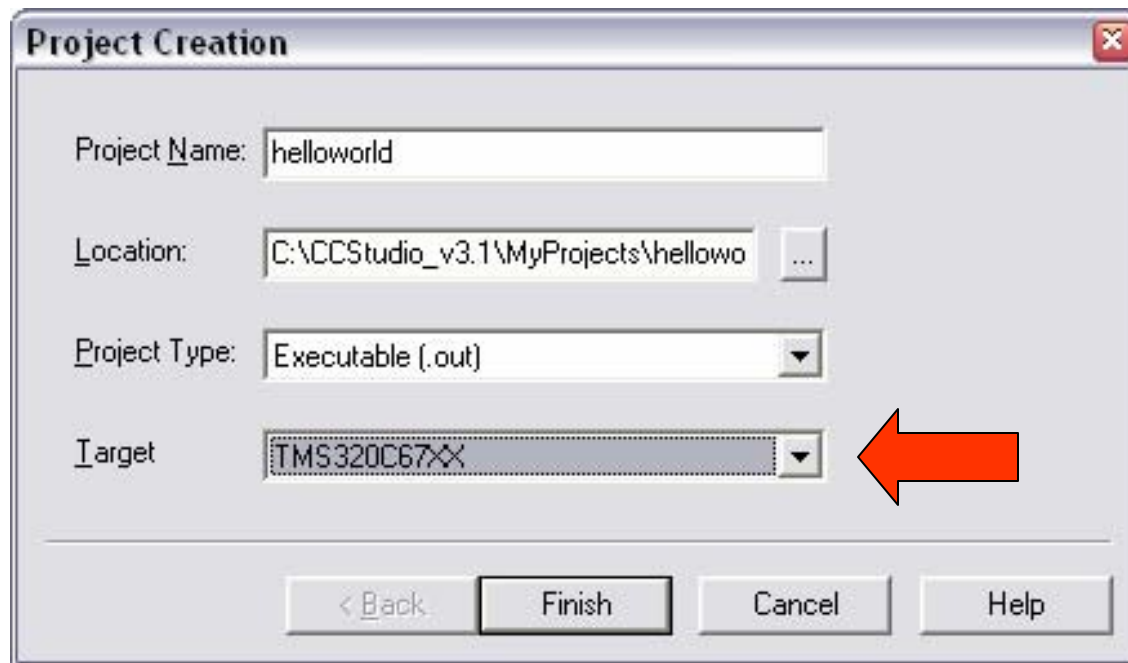
- Open Sin8\_LED project and fix the search path and the memory model (see previous pages). Then build, load, and run it.
  - Press DIP switch 0. You should see LED 0 light up and a 1kHz sinusoid should appear on the left channel of the codec. This is a good test to see if the DSK is working.
- Make 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”



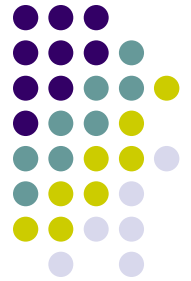


# Creating a New Project (1 of 5)

1. Create new project  
**Project->New**

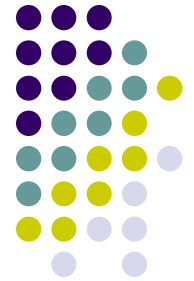


# Creating a New Project (2 of 5)



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)

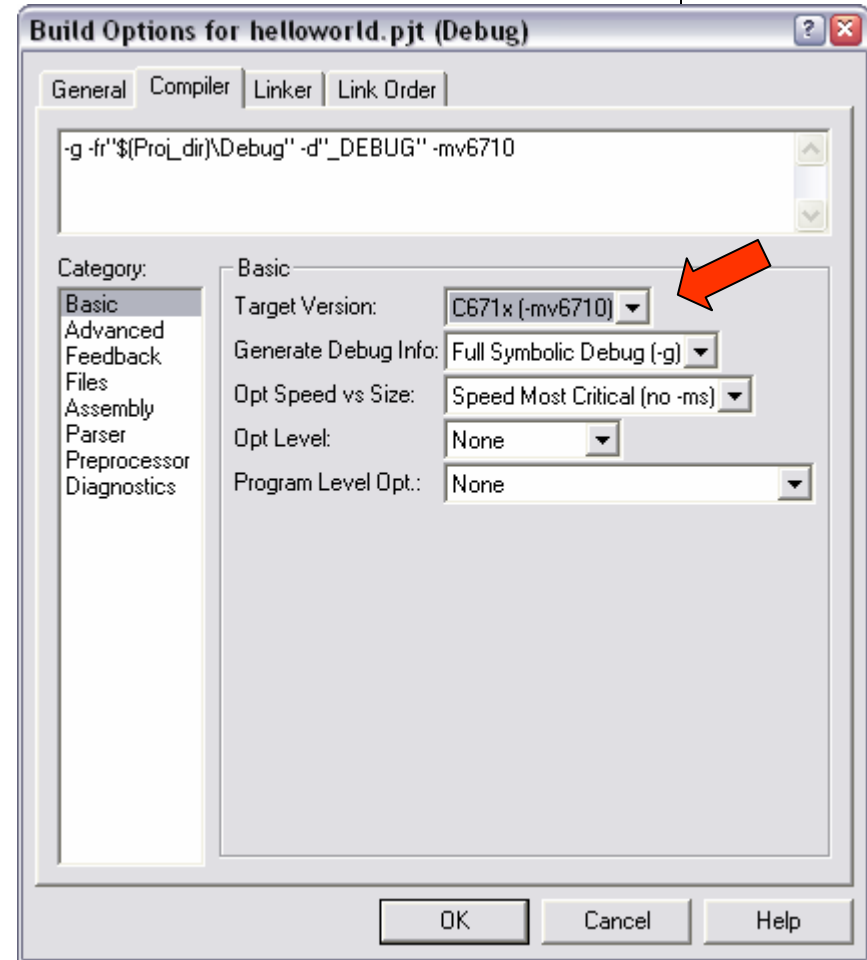
5. Add required support files to project  
**Project->Add Files to Project**
  - a) **myprojects\support\c6713dsk.cmd**  
[linker command file – this or another cmd file is **required**]
  - b) **c6000\cgtools\lib\rts6700.lib**  
[run-time support library functions - **required**]
6. Add optional support 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 level is “none” (this will help with debugging)





# Creating a New Project (5 of 5)

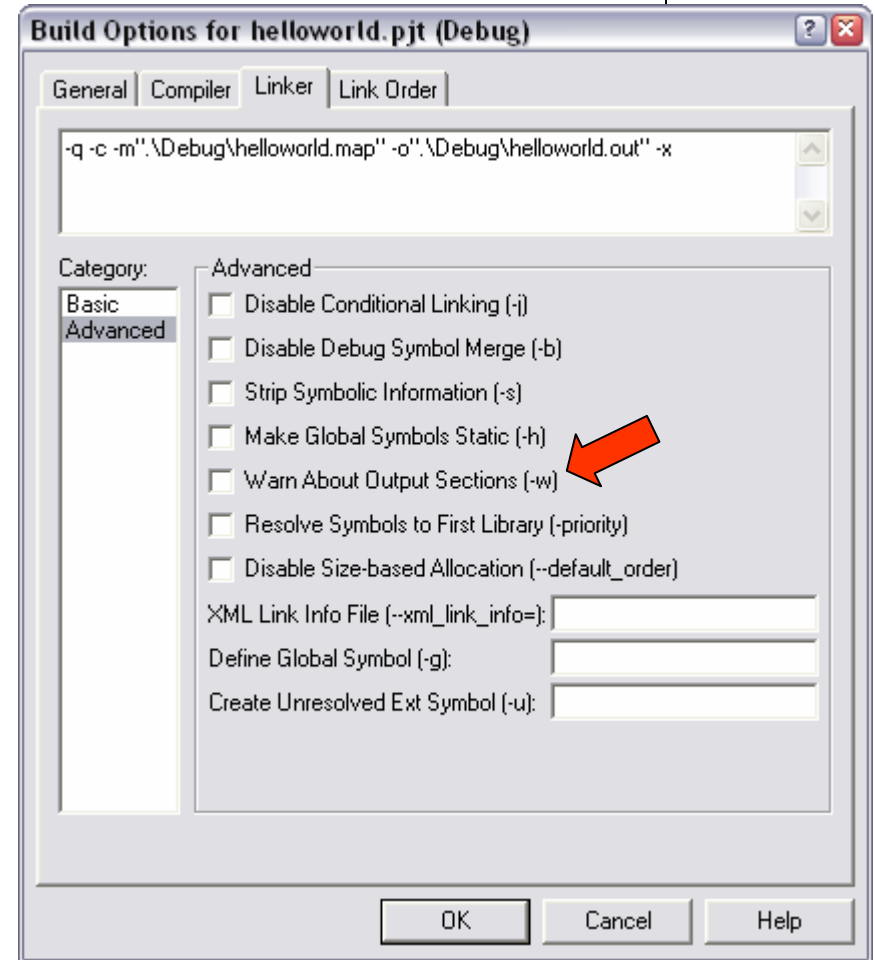
8. Scan all file dependencies to automatically bring all header files and includes into the project:  
**Project -> Scan all file dependencies**
9. Build the project:  
**Project -> Build**
10. If successful, load the .out file to the DSK:  
**File -> Load Program**  
Select the Debug directory. Select 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)



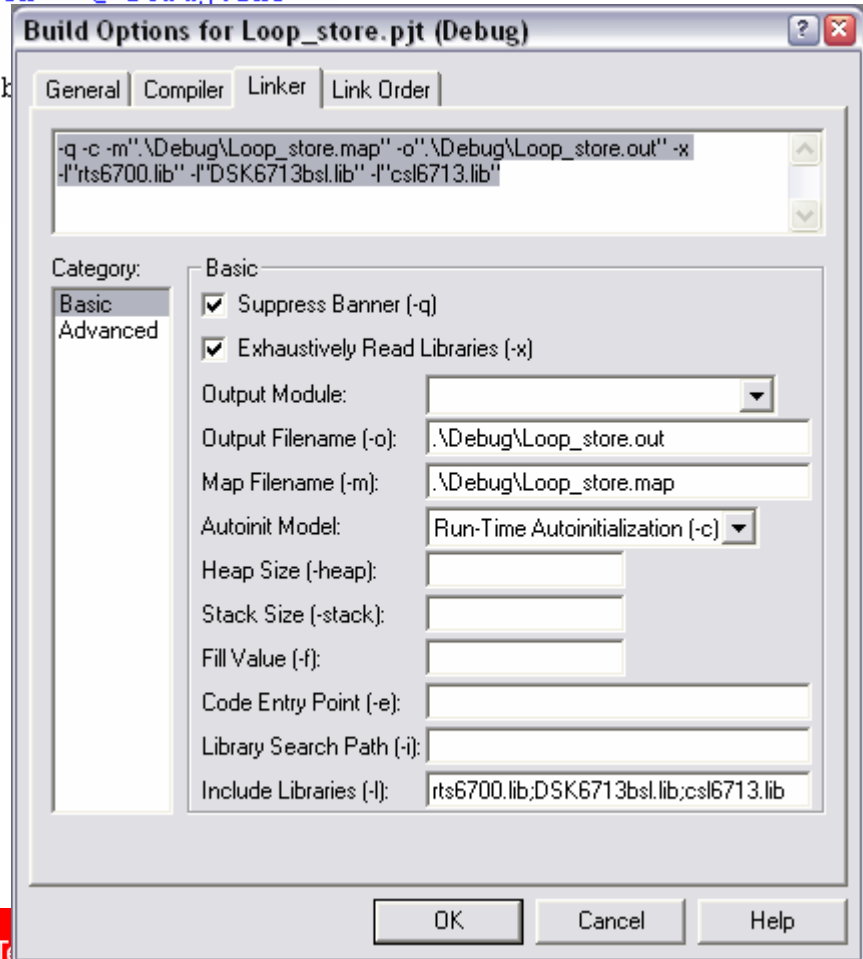
# Tip: Problems finding files during linking



```
[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.
```

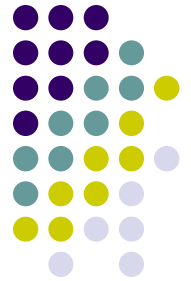


Problem is caused by a bad path for the include libraries in the linker options (Project -> Build Options -> Linker tab)

A fix for this is do 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");

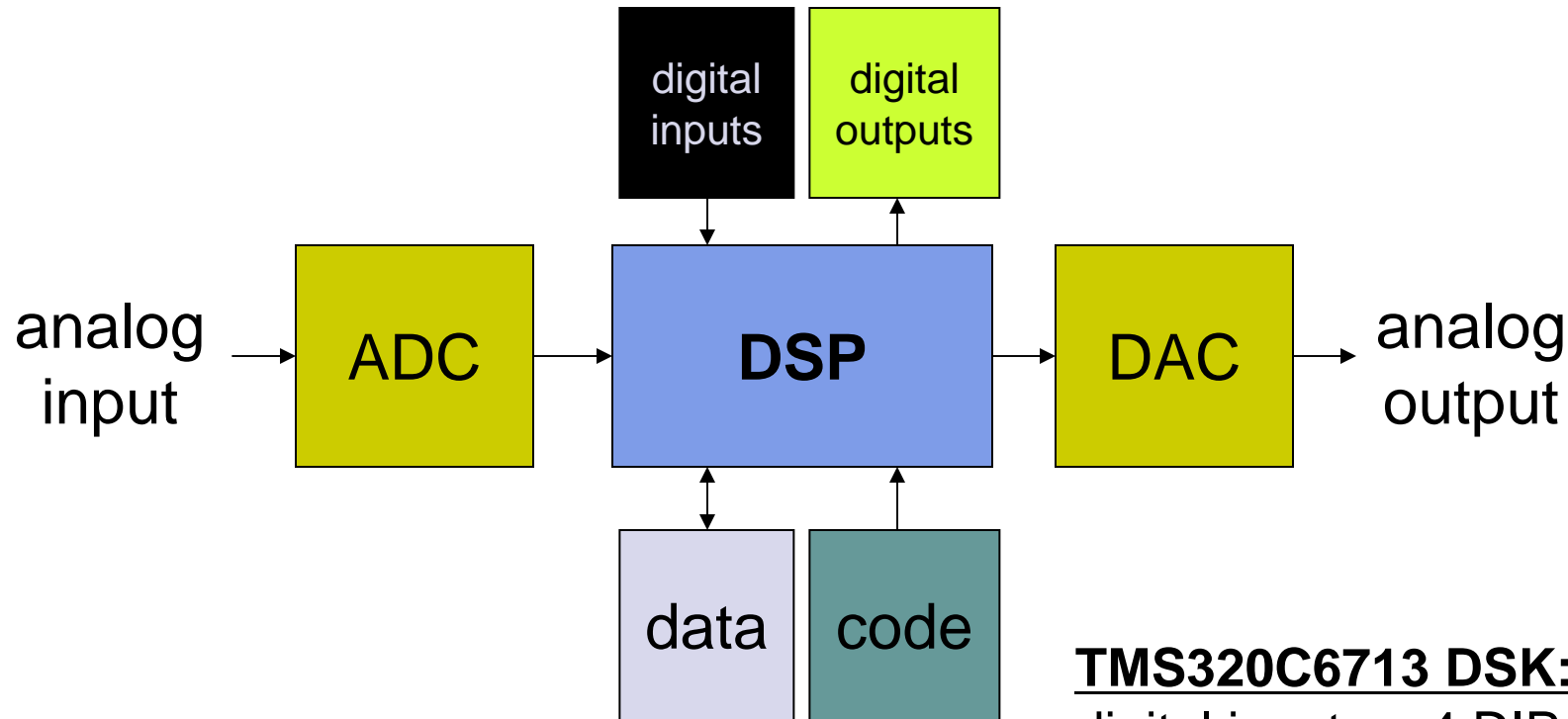
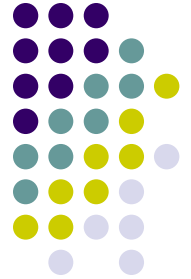
}
```



TEXAS INSTRUMENTS

Technology for Innovators™

# More Interesting Programs: Interfacing with the Real World



**TMS320C6713 DSK:**  
digital inputs = 4 DIP switches  
digital outputs = 4 LEDs  
ADC and DAC = AIC23 codec

# Interfacing with the DIP Switches and LEDs



- 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  $3.$

These functions are provided in **dsk6713bsl.lib**.

Documentation is available in  
**C:\CCStudio\_v3.1\docs\hlp\c6713dsk.hlp**





# Interfacing with the AIC23 Codec

- Determine if you will use **interrupts** or polling
- Initialize the DSK and open the codec
- Set the sampling rate
- Optional configuration
  - Codec
  - Serial ports (McBSP)
- Configure and enable interrupts (if appropriate)
- Input sample(s) from left, right, or both channels
- Output sample(s) to left, right, or both channels

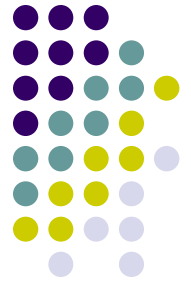




# Polling vs. Interrupts

- Polling
  - Simpler to implement and understand
  - Processing resources wasted by constantly checking codec to see if it is ready for input or output
- Interrupts
  - Slightly more difficult to implement
  - Must set up interrupt vectors to correctly point to interrupt service routine(s) (usually done with a “vectors.asm” file)
  - Much less wasteful of processing resources: codec notifies DSP when ready for input or output





# C6x Interrupt Basics

- 16 interrupt sources (timers, serial ports, ...)
- 12 interrupt events (INT4 to INT15)
- Interrupt events must be mapped to interrupt sources
- Interrupt vectors must be set up. An “interrupt vector” is a special pointer to the start of the “interrupt service routine” (ISR). There are 12 interrupt vectors.
- Interrupt service routine must be set up
  - Save registers
  - Do something useful
  - Restore registers





```
/*
 * This code is written by N. Kehtarnavaz and N. Kim as part of the
 * textbook "Real-Time Digital Signal Processing Based on TMS320C6000"
 * Modified by D. Richard Brown III on 10-Oct-2006
 */
```

```
#define CHIP_6713 1
```

```
#include <stdio.h>
#include <c6x.h>
#include <csl.h>
#include <csl_mcbasp.h>
#include <csl_irq.h>
```

```
#include "dsk6713.h"
#include "dsk6713_aic23.h"
```

```
DSK6713_AIC23_CodecHandle hCodec; // Codec handle
DSK6713_AIC23_Config config = DSK6713_AIC23_DEFAULTCONFIG; // Codec configuration with default settings
```

```
Interrupt void serialPortRcvISR(void); // ISR function prototype
```

```
void main()
{
```

```
    DSK6713_init(); // Initialize the board support library
    hCodec = DSK6713_AIC23_openCodec(0, &config); // Open the codec
    DSK6713_AIC23_setFreq(hCodec, DSK6713_AIC23_FREQ_48KHZ); // set the sampling rate
```

```
    // Configure buffered serial ports for 32 bit operation (L+R in one read/write)
    MCBSP_FSETS(SPCR1, RINTM, FRM);
    MCBSP_FSETS(SPCR1, XINTM, FRM);
    MCBSP_FSETS(RCR1, RWDLEN1, 32BIT);
    MCBSP_FSETS(XCR1, XWDLEN1, 32BIT);
```

```
    // 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)
    {
    }
}
```

## Initialization of Interrupt Interface with AIC23 Codec



TEXAS INSTRUMENTS

Technology for Innovators™



# Stereo read/write ISR

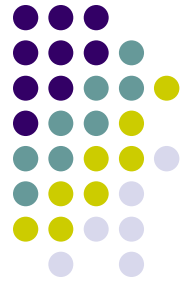
```
interrupt void serialPortRcvISR()  
{  
    Uint32 temp;  
  
    temp = MCBSP_read(DSK6713_AIC23_DATAHANDLE); // read L+R channels  
    MCBSP_write(DSK6713_AIC23_DATAHANDLE, temp); // write L+R channels  
}
```

**Least significant 16 bits of temp = Left channel**  
**Most significant 16 bits of temp = Right channel**

***Note that ISR is only called when INT15 occurs.***



# Opening the Codec and Setting the Sampling Rate

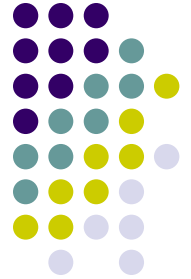


```
DSK6713_init();           // Initialize the DSK
hCodec = DSK6713_AIC23_openCodec(0, &config);
DSK6713_AIC23_setFreq(hCodec, freq);
```

Default configuration: variable “config” declared in `dsk6713_aic23.h`  
Frequency definitions are in `dsk6713_aic.h`

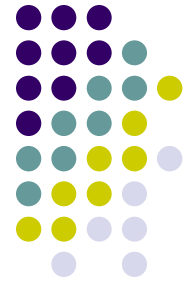
```
/* Frequency Definitions */
#define DSK6713_AIC23_FREQ_8KHZ      1
#define DSK6713_AIC23_FREQ_16KHZ   2
#define DSK6713_AIC23_FREQ_24KHZ   3
#define DSK6713_AIC23_FREQ_32KHZ   4
#define DSK6713_AIC23_FREQ_44KHZ   5
#define DSK6713_AIC23_FREQ_48KHZ   6
#define DSK6713_AIC23_FREQ_96KHZ   7
```





# 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 esoteric settings, e.g. byte order, etc.
- For more details, see
  - [dsk6713\\_aic23.h](#)
  - Codec datasheet (TLV320AIC23B)
  - C:\CCStudio\_v3.1\docs\hlp\c6713dsk.hlp



# Interrupt Setup

## In codec.c:

```
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
```

## In vectors.asm:

```
.ref  _c_int00
.ref  _serialPortRcvISR ; refer the address of ISR defined in C program

.sect "vectors"
... (other interrupts here) ...
INT15:
    MVKL .S2 _serialPortRcvISR, B0
    MVKH .S2 _serialPortRcvISR, B0
    B    .S2 B0
    NOP
    NOP
    NOP
    NOP
    NOP
```

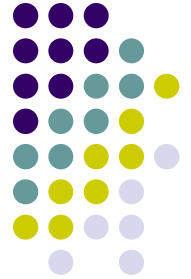




# Some Things to Try

- 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 interrupts to interface to the AIC23.
  - Samples 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



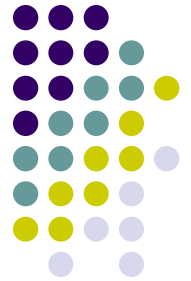
Workshop resumes at 1:30pm...



TEXAS INSTRUMENTS

Technology for Innovators™

# Debugging and Other Useful Features of the CCS IDE

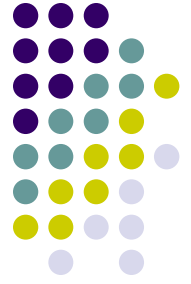


- Breakpoints
- Probe points
- Watch variables
- Plotting arrays of data
- Animation
- General Extension Language (GEL)

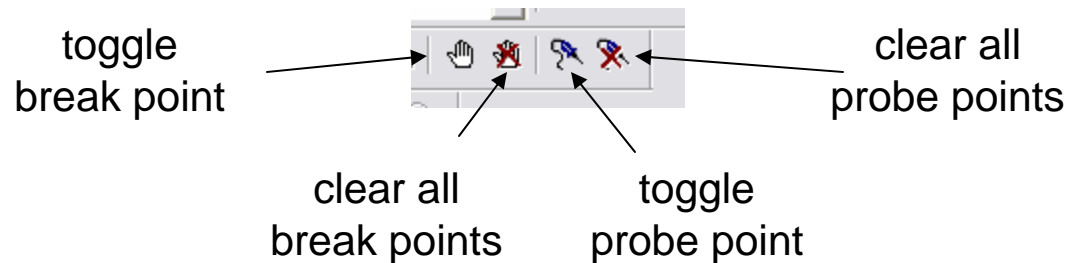


TEXAS INSTRUMENTS

Technology for Innovators™

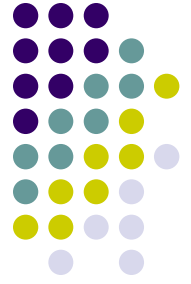


# Breakpoints and Probe Points



```
break point → comm_poll();  
probe point → DSK6713_LED_init();  
DSK6713_DIP_init();  
while(1)  
{  
if(DSK6713_DIP_get(0)==0)  
{  
    //init DSK,codec,McBSP  
    //init LED from BSL  
    //init DIP from BSL  
    //infinite loop  
}  
}  
//=0 if DIP switch #0 pressed
```

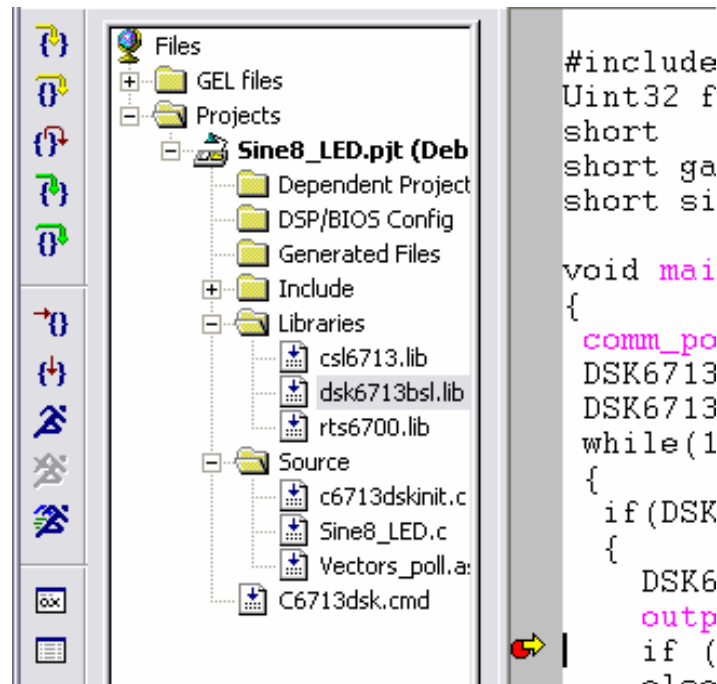
- **Breakpoints:** stop code execution at this point to allow state examination and step-by-step execution.
- **Probe points:** force window updates and/or read/write samples from/to a file at a specific point in your code.



# Breakpoints

- source step into →
- source step over →
- step out →
- ASM step into →
- ASM step over →
  
- run to cursor →
- set program counter to cursor →

“Run to Cursor” is a handy shortcut instead of setting a breakpoint



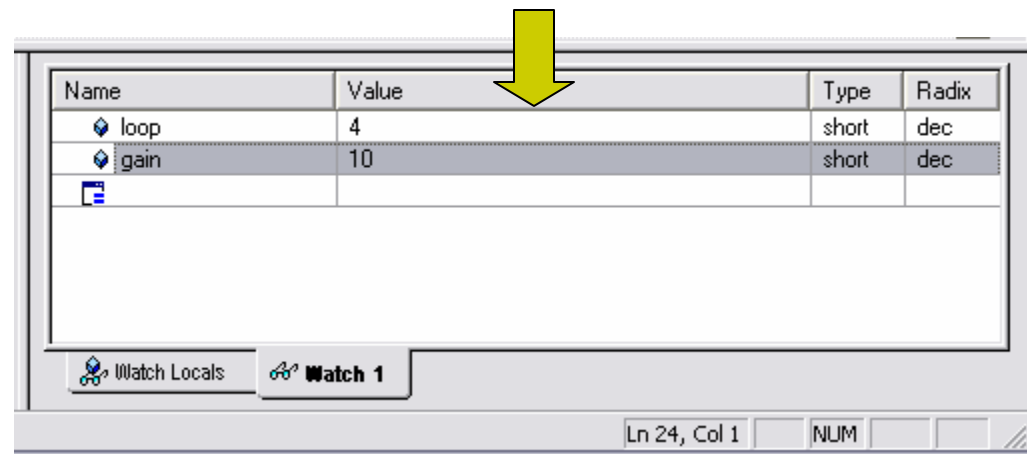
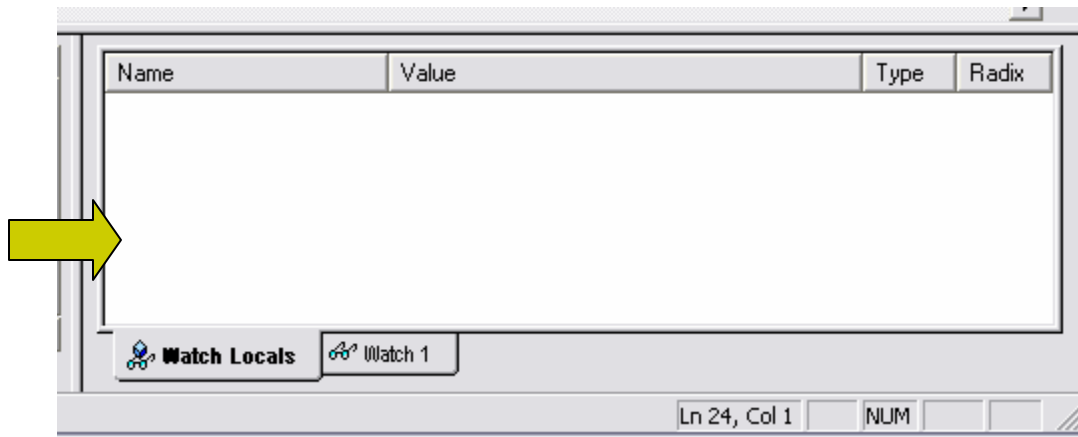
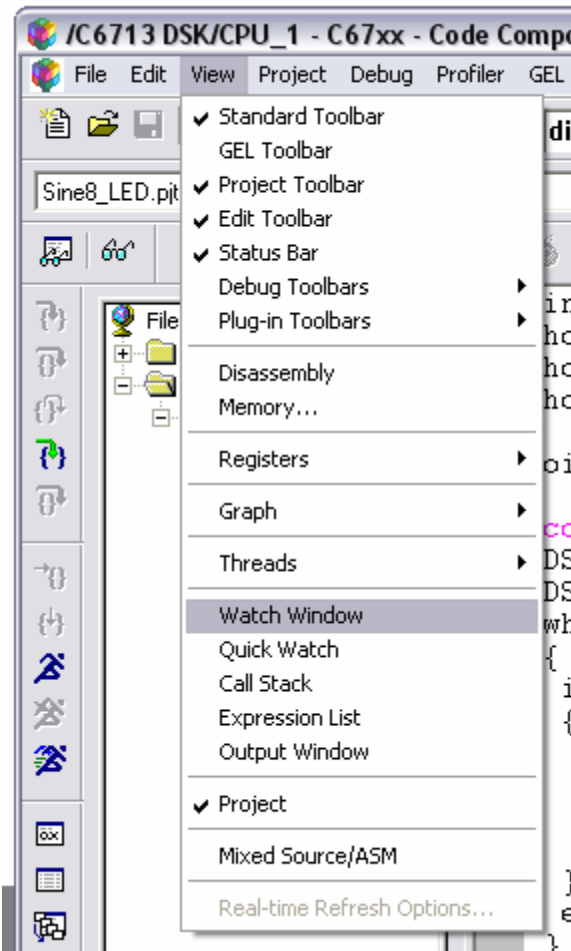


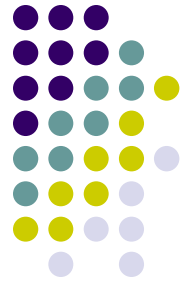
# Probe Points

- 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.



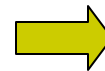


# Plotting Arrays of Data

The screenshot shows the Code Composer Studio interface. The 'View' menu is open, and the 'Graph' option is selected. The background shows a code editor with the following code:

```
dip
int32 fs = DSK6
short loop = 0
short gain = 10;
short sine_table

void main()
{
    if (DSK6713_DIP
    {
        DSK6713_LED_
        output_sampl
        if (loop < 7
        else loop =
    }
    else DSK6713_L
```

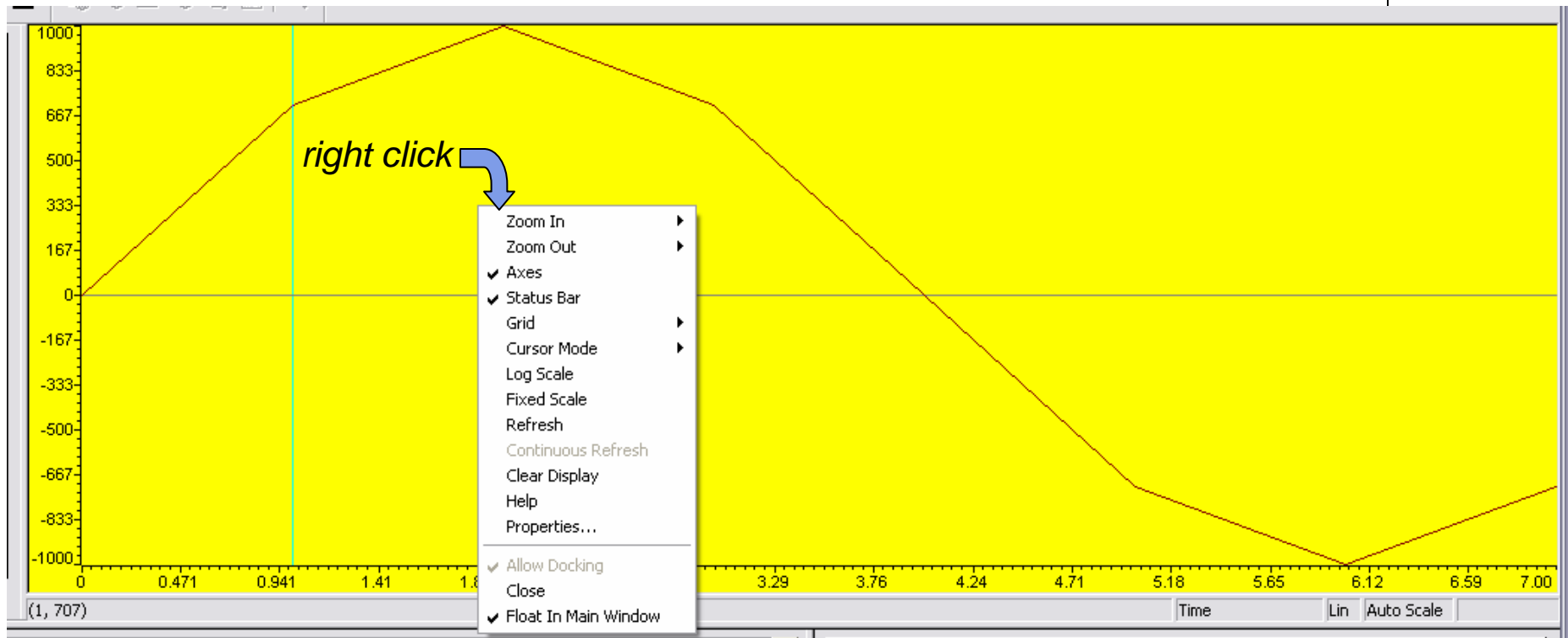


The Graph Property Dialog box is shown with the following settings:

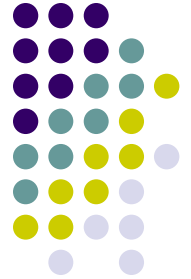
Property	Value
Display Type	Single Time
Graph Title	Graphical Display
Start Address	sine_table
Acquisition Buffer Size	8
Index Increment	1
Display Data Size	8
DSP Data Type	16-bit signed integer
Q-value	0
Sampling Rate (Hz)	8000
Plot Data From	Left to Right
Left-shifted Data Display	Yes
Autoscale	On
DC Value	0
Axes Display	On
Time Display Unit	s
Status Bar Display	On
Magnitude Display Scale	Linear
Data Plot Style	Line
Grid Style	Zero Line
Cursor Mode	Data Cursor



# Plotting Arrays of Data



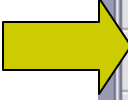
# Animation

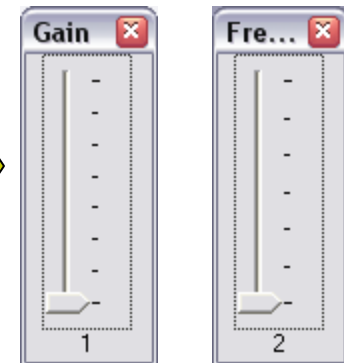
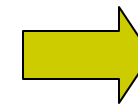
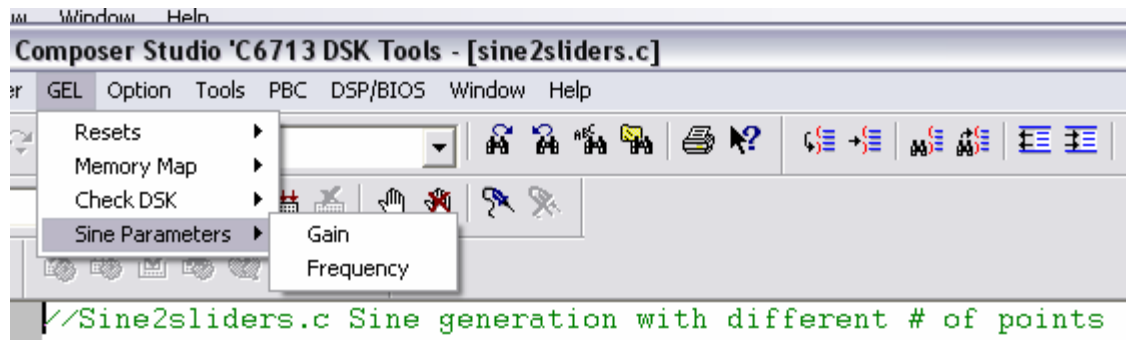
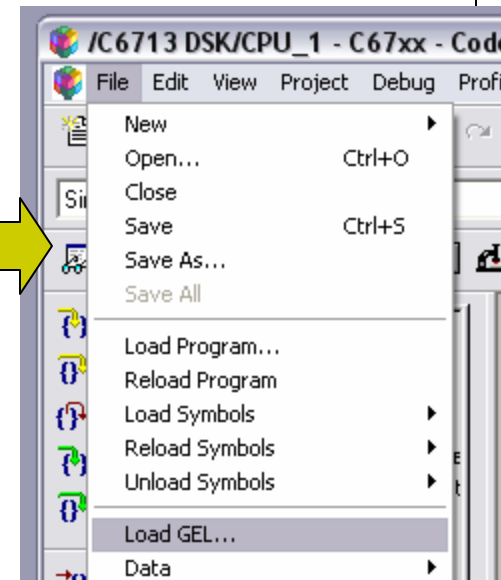


- Runs the program until a breakpoint is encountered.
  - At the breakpoint, execution stops and all windows not connected to any Probe Points are updated.
  - Program execution then automatically resumes
  - Useful for updating graphical displays
  - Note: Animation may cause problems with real-time 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](#)





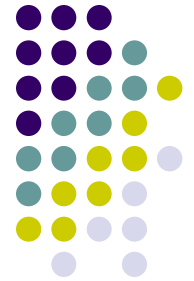
# General Extension Language

- 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 \(GEL\)](#)



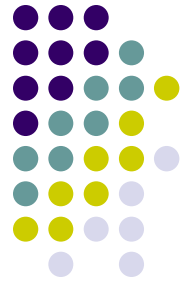


# Some Things to Try

- 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

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

$\mathbf{x}$ =input,  $\mathbf{y}$ =output,  $\mathbf{h}$ =filter coefficients,  $\mathbf{M}$ =# of filter coefficients



TEXAS INSTRUMENTS

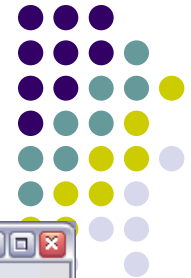
Technology for Innovators™



# 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 **CCS**  
(input samples, output samples, result of multiplies, result of additions)
6. Write code to realize filter
7. Test filter and compare to theoretical expectations

# Designing FIR Filters



>> fdatool

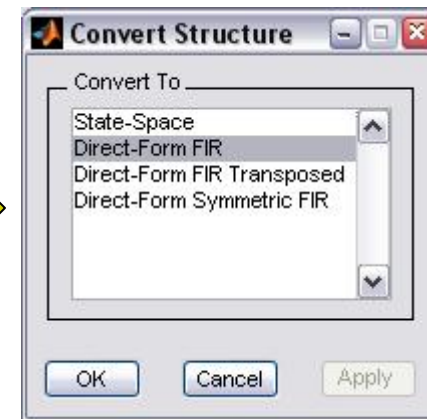
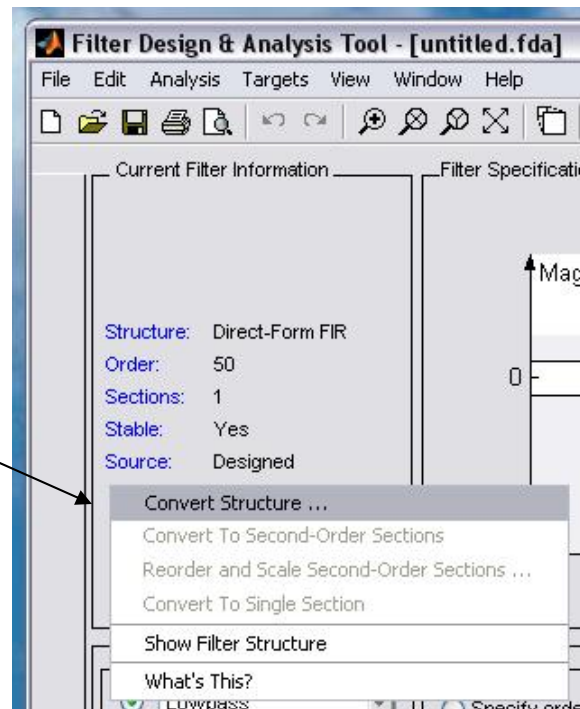




# Filter Realization Structures

- 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

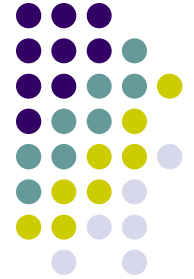


The screenshot shows the 'Filter Design & Analysis Tool' window. The 'Current Filter Information' panel on the left lists: Structure: Direct-Form FIR, Order: 50, Sections: 1, Stable: Yes, Source: Designed. The 'Filter Specifications' panel on the right contains a magnitude response plot with 'Mag. (dB)' on the y-axis and 'f (Hz)' on the x-axis. The plot shows a passband from 0 to  $F_{pass}$  and a stopband from  $F_{stop}$  to  $F_s/2$ . The passband ripple is labeled  $A_{pass}$  and the stopband attenuation is labeled  $A_{stop}$ . Below the plot are two buttons: 'Store Filter ...' and 'Filter Manager ...'.

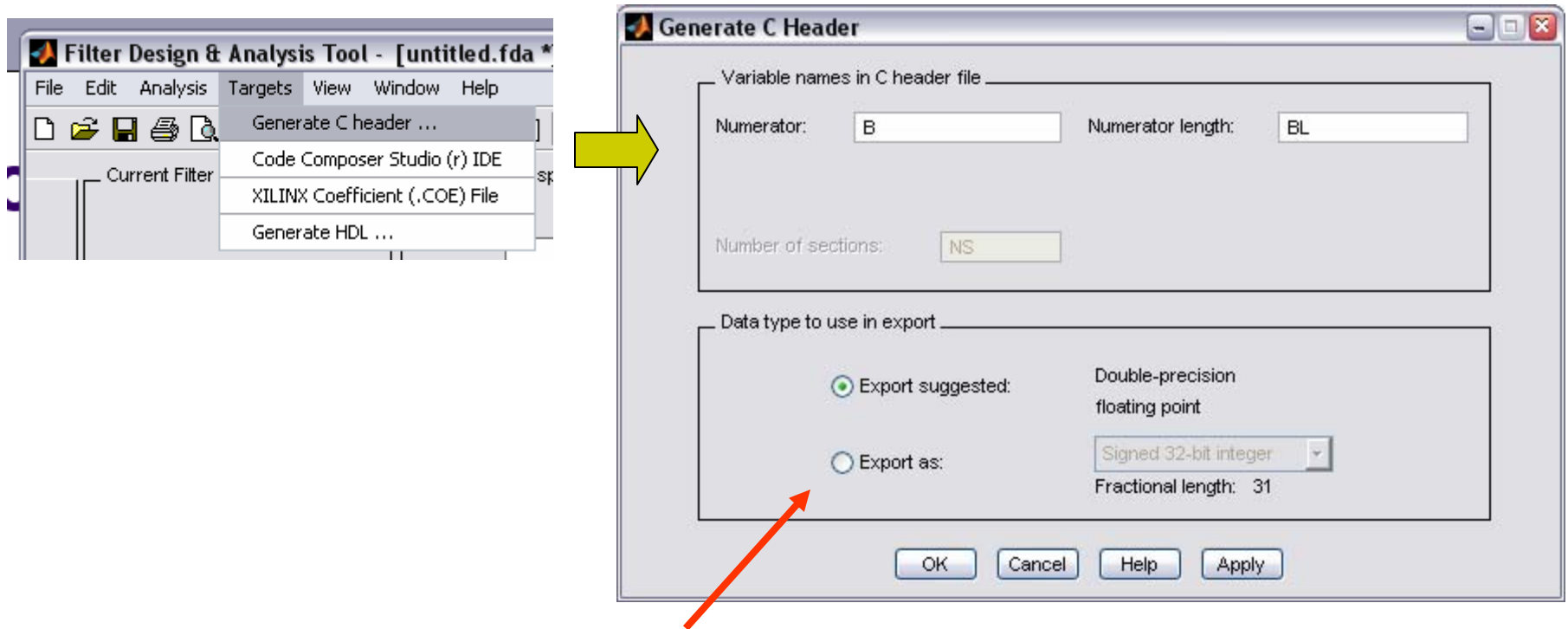
The bottom section of the tool is divided into four panels:

- Response Type:** Lowpass (selected), Highpass, Bandpass, Bandstop, Differentiator.
- Filter Order:** Specify order: 10, Minimum order (selected).
- Options:** Density Factor: 20.
- Frequency Specifications:** Units: Hz,  $F_s$ : 48000,  $F_{pass}$ : 9600,  $F_{stop}$ : 12000.
- Magnitude Specifications:** Units: dB,  $A_{pass}$ : 1,  $A_{stop}$ : 80.

A red arrow points from the text 'set up filter and press' to the 'Design Filter' button at the bottom right of the tool.

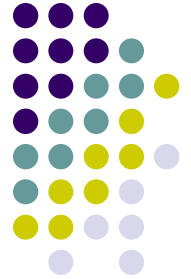


# Make Coefficient File For CCS



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)
 */

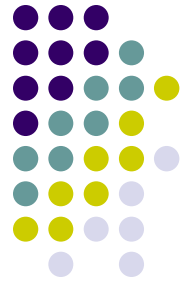
/* General type conversion for MATLAB generated C-code */
#include "tmwtypes.h"
/*
 * Expected path to tmwtypes.h
 * C:\MATLAB7\extern\include\tmwtypes.h
 */
const int BL = 9;
const real64_T B[9] = {
    0.02588139692752, 0.08678803067191, 0.1518399865268, 0.2017873498839,
    0.2205226777929, 0.2017873498839,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

 *Can edit these to agree with your code.*



TEXAS INSTRUMENTS

Technology for Innovators™



# 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

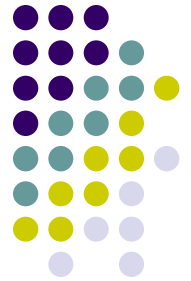
```
interrupt void serialPortRcvISR()
{
    union {Uint32 combo; short channel[2];} temp;
    int i = 0;
    float result = 0.0;

    temp.combo = MCBSP_read(DSK6713_AIC23_DATAHANDLE);

    // Update array samples (move data)
    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;
    MCBSP_write(DSK6713_AIC23_DATAHANDLE, temp.combo);
}
```

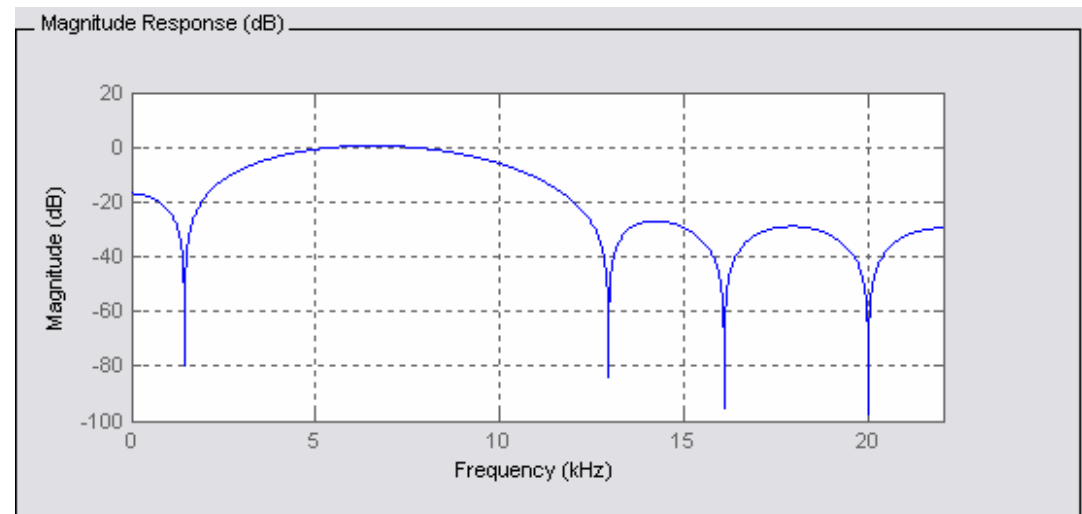
***Note that all math here is floating point.  
Filter coefficients are also assumed to be floating point.***





# Some Things to Try

- Try creating an FIR filter with the following specs:
  - Bandpass
  - 8<sup>th</sup> 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

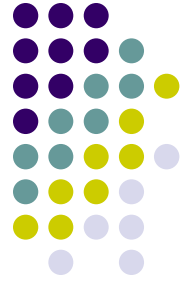
## What you learned today:

- 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](http://www.ti.com/sc/docs/psheets/man_dsp.htm)***

