SECTION 7

DSP HARDWARE

- Microcontrollers, Microprocessors, and Digital Signal Processors (DSPs)
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- DSP Benchmarks
- DSP Evaluation and Development Tools

DSP HARDWARE

SECTION 7 DSP HARDWARE Dan King, Greg Geerling, Ken Waurin, Noam Levine, Jesse Morris, Walt Kester

MICROCONTROLLERS, MICROPROCESSORS, AND DIGITAL SIGNAL PROCESSORS (DSPS)

Computers are extremely capable in two broad areas: (1) *data manipulation*, such as word processing and database management, and (2) *mathematical calculation*, used in science, engineering, and Digital Signal Processing. However, most computers are not optimized to perform *both* functions. In computing applications such as word processing, data must be stored, sorted, compared, moved, etc., and the time to execute a particular instruction is not critical, as long as the program's overall response time to various commands and operations is adequate enough to satisfy the end user. Occasionally, mathematical operations may also be performed, as in a spreadsheet or database program, but speed of execution is generally not the governing factor. In most general purpose computing applications there is no concentrated attempt by software companies to make the code efficient. Application programs are loaded with "features" which require more memory and faster processors with every new release or upgrade.

GENERAL COMPUTING APPLICATIONS

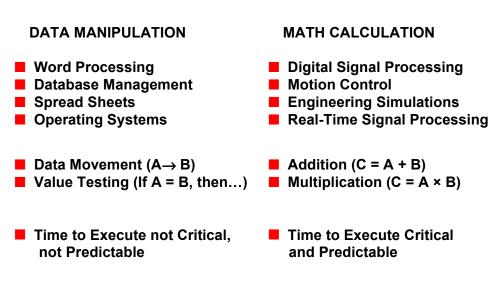


Figure 7.1

On the other hand, digital signal processing applications require that mathematical operations be performed quickly, and the time to execute a given instruction must be known precisely, and it must be predictable. Both code and hardware must be extremely efficient to accomplish this. As has been shown in the last two sections of this book, the most fundamental mathematical operation or *kernel* in all of DSP is

DSP HARDWARE

the sum-of-products (or *dot-product*). Fast execution of the dot product is critical to fast Fourier transforms (FFTs), real time digital filters, matrix multiplications, graphics pixel manipulation, etc.

Based on this introductory discussion of DSP requirements, it is important to understand the differences between *microcontrollers*, *microprocessors*, and *DSPs*. While microcontrollers used in industrial process control applications can perform functions such as multiplication, addition, and division, they are much more suited to applications where I/O capability and control is more important than speed. Microcontrollers such as the 8051-family typically contain a CPU, RAM, ROM, serial/parallel interfaces, timers, and interrupt circuitry. The MicroConverter[™] series from Analog Devices contains not only the 8051 core but also high performance ADC and DAC functions along with flash memory.

MICROCONTROLLERS, MICROPROCESSORS, AND DIGITAL SIGNAL PROCESSORS (DSPs)

Microcontrollers:

- CPU, RAM, ROM, Serial/Parallel Interface, Timer, Interrupt Circuitry
- Well Suited for Toasters as well Industrial Process Control
- Speed is not Generally a Requirement!
- Compact Instruction Sets
- Example: 8051, 68HC11, PIC
- Microprocessors:
 - Single Chip CPU Requires Additional External Circuitry
 - RISC: Reduced Instruction Set Computer
 - CISC: Complex Instruction Set Computer
 - Example: Pentium-Series, PowerPC, MIPS
- Digital Signal Processors (DSPs):
 - RAM, ROM, Serial/Parallel Interface, Interrupt Circuitry
 - CPU Optimized for Fast Repetitive Math for Real Time Processing
 - Example: ADSP-21XX, ADSP-21K

Figure 7.2

Microprocessors, such as the Pentium-series from Intel, are basically single-chip CPUs which require additional circuitry to make up the total computing function. Microprocessor instruction sets can be either complex-instruction-set computer (CISC) or reduced-instruction-set computer (RISC). The complex-instruction-set computer (CISC) includes instructions for basic processor operations, plus single instructions that are highly sophisticated; for example, to evaluate a high-order polynomial. But CISC has a price: many of the instructions execute via microcode in the CPU and require numerous clock cycles plus silicon real estate for code storage memory. In contrast, the reduced-instruction-set computer (RISC) recognizes that, in many applications, basic instructions such as LOAD and STORE - with simple addressing schemes - are used much more frequently than the advanced instructions, and should not incur an execution penalty. These simpler instructions are hardwired in the CPU logic to execute in a single clock cycle, reducing execution time and CPU complexity.

Although the RISC approach offers many advantages in general purpose computing, it is not well suited to DSP. For example, most RISCs do not support singleinstruction multiplication, a very common and repetitive operation in DSP. The DSP is optimized to accomplish these tasks fast enough to maintain real-time operation in the context of the application. This requires single-cycle arithmetic operations and accumulations.

DSP REQUIREMENTS

The most fundamental mathematical operation in DSP is shown in Figure 7.3: the sum of products (dot product). It is common to digital filters, FFTs, and many other DSP applications. A Digital Signal Processor (DSP) is optimized to perform repetitive mathematical operations such as the dot product. There are five basic requirements of a DSP to optimize this performance: *fast arithmetic, extended precision, dual operand fetch, circular buffering,* and *zero-overhead looping.*

THE MOST FUNDAMENTAL MATHEMATICAL OPERATION IN DSP: THE SUM OF PRODUCTS

- $y(n) = h(0) \cdot x(n) + h(1) \cdot x(n-1) + ... + h(N-1) \cdot x(n-N)$
 - For Example: Digital Filtering
 - Multiply Data Sample Times Filter Coefficient (Twiddle Factor for FFTs)
 - Add Contents to Accumulator
 - Repeat N times
 - **DSP** Requirements:
 - Fast Multiply-Accumulates
 - Extended Precision (Accumulator Register)
 - Dual Operand Fetch
 - Circular Buffering
 - Zero-Overhead Looping
 - In One Instruction Cycle Using ADSP-21XX Core :
 - Fetch Data Sample from Data Memory
 - Fetch Coefficient from Program Memory
 - Perform Multiply-Accumulate
 - Update Pointers

Figure 7.3

DSP HARDWARE

FAST ARITHMETIC

Fast arithmetic is the simplest of these requirements to understand. Since real-time DSP applications are driven by performance, the multiply-accumulate or MAC time is a basic requirement; faster MACs mean potentially higher bandwidth. It is critical to remember that MAC time alone does not define DSP performance. This often forgotten fact leads to an inadequate measure of processor performance by simply examining its MIPS (million instructions per second) rating. Since most DSP and DSP-like architectures feature MACs that can execute an instruction every cycle, most processors are given a MIPS rating equal to its MAC throughput. This does not necessarily account for the other factors that can degrade a processors overall performance in real-world applications. The other four criteria can wipe out MAC gains if they are not satisfied.

In addition to the requirement for fast arithmetic, a DSP should be able to support other general purpose math functions and should therefore have an appropriate arithmetic logic unit (ALU) and a programmable shifter function for bit manipulation.

EXTENDED PRECISION

Apart from the obvious need for fast multiplication and addition (MAC), there is also a requirement for extended precision in the accumulator register. For example, when two 16-bit words are multiplied, the result is a 32-bit word. The Analog Devices ADSP-21xx 16-bit fixed-point core architecture has an internal 40-bit accumulator which provides a high degree of overflow protection. While floatingpoint DSPs eliminate most of the problems associated with precision and overflow, fixed-point processors are still popular for many applications, and therefore overflow, underflow, and data scaling issues must be dealt with properly.

DUAL OPERAND FETCH

Regardless of the nature of a processor, performance limitations are generally based on bus bandwidth. In the case of general purpose microprocessors or microcontrollers, code is dominated by single memory fetch instructions, usually addressed as a base plus offset value. This leads architects to embed fixed data into the instruction set so that this class of memory access is fast and memory efficient. DSPs, on the other hand, are dominated by instructions requiring two independent memory fetches. This is driven by the basic form of the convolution (kernel or dot product) $\Sigma h(i)x(i)$. The goal of fast dual operand fetches is to keep the MAC fully loaded. We saw in the discussion on MACs that the performance of a DSP is first limited by MAC time. Assuming an adequate MAC cycle time, two data values need to be supplied at the same rate; increases in operand fetch cycle time will result in corresponding increases in MAC cycle time. Ideally, the operand fetches occur simultaneously with the MAC instruction so that the combination of the MAC and memory addressing occurs in one cycle.

Dual operand fetch is implemented in DSPs by providing separate buses for program memory data and data memory data. In addition, separate program memory address and data memory address buses are also provided. The MAC can therefore receive inputs from each data bus simultaneously. This architecture is often referred to as the Harvard Architecture.

CIRCULAR BUFFERING

If we examine the kernel equation more carefully, the advantages of circular buffering in DSP applications become apparent. A Finite Impulse Response (FIR) filter is used to demonstrate the point. First, coefficients or tap values for FIR filters are periodic in nature. Second, the FIR filter uses the newest real-world signal value and discards the oldest value when calculating each output.

In the series of FIR filter equations, the N coefficient locations are always accessed sequentially from h(0) to h(N-1). The associated data points circulate through the memory as follows: new samples are stored replacing the oldest data each time a filter output is computed. A fixed boundary RAM can be used to achieve this circulating buffer effect. The oldest data sample is replaced by the newest with each convolution. A "time history" of the N most recent samples is kept in RAM.

This delay line can be implemented in fixed boundary RAM in a DSP chip if new data values are written into memory, overwriting the oldest value. To facilitate memory addressing, old data values are read from memory starting with the value one location after the value that was just written. In a 4-tap FIR filter, for example, x(4) is written into memory location 0, and data values are then read from locations 1, 2, 3, and 0. This example can be expanded to accommodate any number of taps. By addressing data memory locations in this manner, the address generator need only supply sequential addresses regardless of whether the operation is a memory read or write. This data memory buffer is called *circular* because when the last location is reached, the memory pointer must be reset to the beginning of the buffer.

The coefficients are fetched simultaneously with the data. Due to the addressing scheme chosen, the oldest data sample is fetched first. Therefore, the last coefficient must be fetched first. The coefficients can be stored backwards in memory: h(N-1) is the first location, and h(0) is the last, with the address generator providing incremental addresses. Alternatively, coefficients can be stored in a normal manner with the accessing of coefficients starting at the end of the buffer, and the address generator being decremented.

This allows direct support of the FIR filter unit delay taps without software overhead. These data characteristics are DSP algorithm-specific and must be supported in hardware to achieve the best DSP performance. Implementing circular buffers in hardware allows buffer parameters (i.e. start, length, etc.) to be set up outside of the core instruction loop. This eliminates the need for extra instructions within the loop body. Lack of a hardware implementation for circular buffering can significantly impact MAC performance.

ZERO OVERHEAD LOOPING

Zero overhead looping is required by the repetitive nature of the kernel equation. The multiply-accumulate function and the data fetches required are repeated N times every time the kernel function is calculated. Traditional microprocessors

DSP HARDWARE

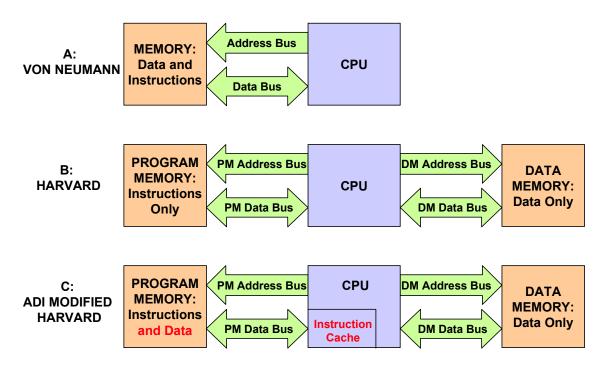
implement loops that have one instruction execution time or more of overhead associated with repeating the loop. Analog Devices' DSP architectures provide hardware support that eliminates the need for looping instructions within the loop body. For true DSP architectures, the difference of zero overhead body looping and programmed looping can easily exceed 20% cycle time.

SUMMARY

Any processor can accomplish any software task, given enough time. However, DSPs are optimized for the unique computational requirements of real-time, realworld signal processing. Traditional computers are better suited for tasks that can be performed in non-real-time. In the following section, we will examine the architecture of a high-performance 16-bit fixed-point DSP Microcomputer, the Analog Devices' ADSP-21xx-family.

ADSP-21XX 16-BIT FIXED-POINT DSP CORE

Traditional microprocessors use the *Von Neumann architecture* (named after the American mathematician John Von Neumann) as shown in Figure 7.4A. The Von Neumann architecture consists of a single memory which contains data and instructions and a single bus for transferring data and instructions into and out of the CPU. Multiplying two numbers requires at least three cycles: two cycles are required to transfer the two numbers into the CPU, and one cycle to transfer the instruction. This architecture is satisfactory when all the required tasks can be executed serially. In fact, most general purpose computers today use the Von Neumann architecture.



MICROPROCESSOR ARCHITECTURES

Figure 7.4

For faster processing, however, the *Harvard architecture* shown in Figure 7.4B is more suitable. This is named for the work done at Harvard University under the leadership of Howard Aiken. Data and program instructions each have separate memories and buses as shown. Since the buses operate independently, program instructions and data can be fetched at the same time, thereby improving the speed over the single bus Von Neumann design. In order to perform a single FIR filter multiply-accumulate, an instruction is fetched from the program memory, and during the same cycle, a coefficient can be fetched from data memory. A second cycle is required to fetch the data word from the data memory.

Figure 7.4C illustrates Analog Devices' modified Harvard architecture where instructions and data are allowed in the program memory. For example, in the case of a digital filter, the coefficients are stored in the program memory, and the data samples in the data memory. A coefficient and a data sample can thus be fetched in a single cycle. In addition to fetching the coefficient from program memory and a data sample from data memory, an instruction must also be fetched from program memory. Analog Devices' DSPs handle this in one of two ways. In the first method, the program memory is accessed twice (double pumped) in an instruction cycle. The ADSP-218x series uses this method. In the second method, a program memory cache is provided. If an algorithm requires dual data fetches, the programmer places one buffer in program memory and the other in data memory. The first time the processor executes an instruction, there is a one-cycle stall because it must fetch the instruction and the coefficient over the program memory data bus. However, whenever this conflict occurs, the DSP "caches" the instruction in a cache memory. The next time this instruction is required, the program sequencer obtains it from the cache, while the coefficient is obtained over the program memory data bus. The cache method if used in the ADSP-219x family as well as in the SHARC family.

DIGITAL FILTER EXAMPLE

Now that the fundamental architecture of the ADSP-21xx family has been presented, a simple FIR filter design will illustrate the ease of programming the family. Pseudocode for an FIR filter design is shown in Figure 7.5. For Analog Devices' DSPs, all operations within the filter loop are completed in one instruction *cycle*, thereby greatly increasing efficiency. Extra instructions are not required to repeat the loop. This is referred to as zero-overhead looping. The actual FIR filter assembly code for the ADSP-21xx family of fixed point DSPs is shown in Figure 7.6. The arrows in the diagram point to the actual executable instructions (7 lines), the rest of the code are simply comments added for clarification. The first instruction (labeled *fir*:) sets up the computation by clearing the MR register and loading the MX0 and MY0 registers with the first data and coefficient values from data and program memory. The multiply-accumulate with dual data fetch in the *convolution* loop is then executed N-1 times in N-1 cycles to compute the sum of the first N-1products. The final multiply-accumulate instruction is performed with the rounding mode enabled to round the result to the upper 24 bits of the MR register. The MR1 register is then conditionally saturated to its most positive or negative value based on the status of the overflow flag contained in the MV register. In this manner, results are accumulated to the full 40-bit precision of the MR register, with saturation of the output only if the final result overflowed beyond the least significant 32 bits of the MR register.

PSEUDOCODE FOR FIR FILTER PROGRAM USING A DSP WITH CIRCULAR BUFFERING

- 1. Obtain sample from ADC (typically interrupt driven)
- 2. Move sample into input signal's circular buffer
- 3. Update the pointer for the input signal's circular buffer
- 4. Zero the accumulator
- 5. Implement filter (control the loop through each of the coefficients)
 - 6. Fetch the coefficient from the coefficient's circular buffer
 - 7. Update the pointer for the coefficient's circular buffer
 - 8. Fetch the sample from the input signal's circular buffer
 - 9. Update the pointer for the input signal's circular buffer
 - 10. Multiply the coefficient by the sample
 - 11. Add the product to the accumulator

12. Move the filtered sample to the DAC

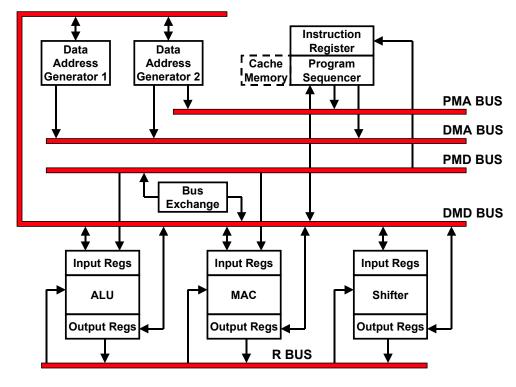
```
ADSP21xx Example code:
CNTR = N-1;
DO convolution UNTIL CE;
convolution:
    MR = MR + MX0 * MY0(SS), MX0 = DM(I0,M1), MY0 = PM(I4,M5);
```

```
Figure 7.5
```

ADSP-21XX FIR FILTER ASSEMBLY CODE (SINGLE PRECISION)

	.MODULE	fir_sub;					
	{	FIR Filter Subroutine					
		Calling Parameters					
		IO> Oldest input data value in delay line I4> Beginning of filter coefficient table LO = Filter length (N)					
		L4 = Filter length (N)					
		M1,M5 = 1					
		CNTR = Filter length - 1 (N-1)					
		Return Values					
		MR1 = Sum of products (rounded and saturated)					
		IO> Oldest input data value in delay line					
		I4> Beginning of filter coefficient table					
		Altered Registers					
		MX0,MY0,MR					
		Computation Time					
		(N - 1) + 6 cycles = $N + 5$ cycles					
		All coefficients are assumed to be in 1.15 format. $\}$					
	.ENTRY	fir;					
\rightarrow	fir:	<pre>MR=0, MX0=DM(I0,M1), MY0=PM(I4,M5);</pre>					
\rightarrow		CNTR = N-1;					
\rightarrow		DO convolution UNTIL CE;					
	convolution:	<pre>MR=MR+MX0*MY0(SS), MX0=DM(I0,M1), MY0=PM(I4,M5);</pre>					
		MR=MR+MX0*MY0(RND);					
\rightarrow		IF MV SAT MR;					
\rightarrow		RTS;					
	.ENDMOD;						

The ADSP-21xx family architecture (Figure 7.7) is optimized for digital signal processing and other high-speed numeric processing applications. This family of DSPs combine the complete ADSP-2100 core architecture (three computational units, data address generators, and a program sequencer) with two serial ports, a programmable timer, extensive interrupt capabilities and on-board program and data memory RAM. ROM-based versions are also available.



ADSP-21XX CORE ARCHITECTURE

Figure 7.7

The ADSP-21xx's flexible architecture and comprehensive instruction set support a high degree of operational parallelism. In one cycle the ADSP-21xx can generate the next program address, fetch the next instruction, perform one or two data moves, update one or two data address pointers, perform a computational operation, receive and transmit data via the two serial ports, and update a timer.

ADSP-21XX CORE ARCHITECTURE

- Buses
 - Program Memory Address (PMA)
 - Data Memory Address (DMA)
 - Program Memory Data (PMD)
 - Data Memory Data (DMD)
 - Result (R)
- Computational Units
 - Arithmetic Logic Unit (ALU)
 - Multiply-Accumulator (MAC)
 - Shifter
- Data Address Generators
- Program Sequencer
- On-Chip Peripheral Options
 - Program Memory RAM or ROM
 - Data Memory RAM
 - Serial Ports
 - Timer
 - Host Interface Port
 - DMA Port

Figure 7.8

BUSES

The ADSP-21xx processors have five internal buses to ensure efficient data transfer. The program memory address (PMA) and data memory address (DMA) buses are used internally for the addresses associated with program and data memory. The program memory data (PMD) and data memory data (DMD) buses are used for the data associated with the memory spaces. Off chip, the buses are multiplexed into a single external address bus and a single external data bus; the address spaces are selected by the appropriate control signals. The result (R) bus transfers intermediate results directly between the various computational units.

The PMA bus is 14-bits wide allowing direct access of up to 16K words of data. The data memory data (DMD) bus is 16-bits wide. The DMD bus provides a path for the contents of any register in the processor to be transferred to any other register or to any data memory location in a single cycle. The data memory address comes from two sources: an absolute value specified in the instruction code (direct addressing) or the output of a data address generator (indirect addressing). Only indirect addressing is supported for data fetches from program memory.

The program memory data (PMD) bus can also be used to transfer data to and from the computational units through direct paths or via the PMD-DMD bus exchange unit. The PMD-DMD bus exchange unit permits data to be passed from one bus to the other. It contains hardware to overcome the 8-bit width discrepancy between the two buses, when necessary. Program memory can store both instructions and data, permitting the ADSP-21xx to fetch two data operands in a single cycle, one from program memory and one from data memory. The corresponding instruction is obtained directly from program memory by "double pumping" (ADSP-218x series) or from a cache memory (ADSP-219x and SHARC series).

COMPUTATIONAL UNITS (ALU, MAC, SHIFTER)

The processor contains three independent computational units: the arithmetic logic unit (ALU), the multiplier-accumulator (MAC), and the barrel shifter. The computational units process 16-bit data directly and have provisions to support multiprecision computations. The ALU has a carry-in (CI) bit which allows it to support 32-bit arithmetic.

The ALU provides a standard set of arithmetic and logic functions: add, subtract, negate, increment, decrement, absolute value, AND, OR, EXCLUSIVE OR and NOT. Two divide primitives are also provided.

ARITHMETIC LOGIC UNIT (ALU) FEATURES

- Add, Subtract, Negate, Increment, Decrement, Absolute Value, AND, OR, Exclusive OR, NOT
- Bitwise Operators, Constant Operators
- Multi-Precision Math Capabilites
- Divide Primitives
- Saturation Mode for Overflow Support
- Background Registers for Single-Cycle Context Switch
- Example Instructions:
 - IF EQ AR = AX0 + AY0;
 - AF = MR1 XOR AY1;
 - ◆ AR = TGLBIT 7 OF AX1;

Figure 7.9

The MAC performs single-cycle multiply, multiply/add, and multiply/subtract operations. It also contains a 40-bit accumulator which provides 8-bits of overflow in successive additions to ensure that no loss of data occurs; 256 overflows would have to occur before any data is lost. Special instructions are provided for implementing block floating-point scaling of data. A set of background registers is also available in the MAC for interrupt service routines. If after a DSP routine is finished and the MV flag has been set, this means that the register contains a word greater than 32 bits. The register can be "saturated" using the saturation routine which normalizes the 40-bit word to either a negative fullscale or positive fullscale 32-bit word in 1.32 format.

MULTIPLY-ACCUMULATOR (MAC) FEATURES

- Single-Cycle Multiply, Multiply-Add, Multiply-Subtract
- 40-Bit Accumulator for Overflow Protection (219x Adds Second 40-Bit Accumulator)
- Saturation Instruction Performs Single Cycle Overflow Cleanup
- Background Registers for Single-Cycle Context Switch
- **Example MAC Instructions:**
 - MR = MX0 * MY0(US);
 - IF MV SAT MR;
 - $\blacklozenge MR = MR AR * MY1(SS);$
 - MR = MR + MX1 * MY0(RND);
 - IF LT MR = MX0 * MX0(UU);

Figure 7.10

The shifter performs logical and arithmetic shifts, normalization, denormalization, and derive-exponent operations. The shifter can be used to efficiently implement numeric format control including multiword floating-point representations.

SHIFTER FEATURES

- Normalize (Fixed-Point to Floating-Point Conversion)
- Denormalize (Floating-Point to Fixed-Point Conversion)
- Arithmetic and Logical Shifts
- Block Floating Point Support
- Derive Exponent
- Background Registers for Single-Cycle Context Switch
- **Example Shifter Instructions:**

•	SR	=	ASI	IIFT	' SI	вү	-6	(LO)	;		{2	Arithmet	ic	Shi	ft}
•	SR	=	SR	OR	LSH	IFT	SI	ву	3 (HI);	{:	Logical	Shi	lft}	

\$ SR = NORM MR1(LO); {Normalization}

The computational units are arranged side-by-side instead of serially so that the output of any unit may be the input of any unit on the next cycle. The internal result (R) bus directly connects the computational units to make this possible.

DATA ADDRESS GENERATORS AND PROGRAM SEQUENCER

Two dedicated data address generators and a powerful program sequencer ensure efficient use of the computational units. The data address generators (DAGs) provide memory addresses when memory data is transferred to or from the input or output registers. Each DAG keeps track of up to four address pointers. Whenever the pointer is used to access data (indirect addressing), it is post-modified by the value of a specified modify register. A length value may be associated with each pointer to implement automatic modulo addressing for circular buffers. With two independent DAGs, the processor can generate two addresses simultaneously for dual operand fetches.

DAG1 can supply addresses to data memory only; DAG2 can supply addresses to either data memory or program memory. When the appropriate mode bit is set in the mode status register (MSTAT), the output address of DAG1 is bit-reversed before being driven onto the address bus. This feature facilitates addressing in radix-2 FFT algorithms.

DATA ADDRESS GENERATOR FEATURES

- Automatic Linear Addressing and Circular Buffering
- Each DAG Manages Four Pointers
- Supports Dual Operand Fetch
- Bit-Reverser (DAG1) For FFTs
- Background Registers in ADSP-219x
- Example DAG Instructions:
 - AX0 = DM(I0,M3);
 - MODIFY(I1, M2);

```
MR = MR + MX0 * MY0, MX0 = DM(I0,M1), MY0 = PM(I4,M4);
Fetch Data Memory Data Word Located at Address I0, Increment Pointer by M1
Fetch Program Memory Data Word Located at Address I4, Increment Pointer by M4
```

Figure 7.12

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The program sequencer supplies instruction addresses to the program memory. The sequencer is driven by the instruction register which holds the currently executing instruction. The instruction register introduces a single level of pipelining into the program flow. Instructions are fetched and loaded into the instruction register during one processor cycle, and executed during the following cycle while the next instruction is prefetched. To minimize overhead cycles, the sequencer supports conditional jumps, subroutine calls and returns in a single cycle. With an internal loop counter and loop stack, the processor executes looped code with zero overhead. No explicit jump instructions are required to loop. The sequencer also efficiently processes interrupts with its interrupt controller for fast interrupt response with minimum latency. When an interrupt occurs, it causes a jump to a known specified location in memory. Short interrupt service routines can be coded in place. For interrupt service routines with more than four instructions, program control is transferred to the service routine by means of a JUMP instruction placed at the interrupt vector location.

PROGRAM SEQUENCER FEATURES

- Generates Next Instruction Address
- Low-Latency Interrupt Handling
- Hardware Stacks

Single-Cycle Conditional Branch (218x)

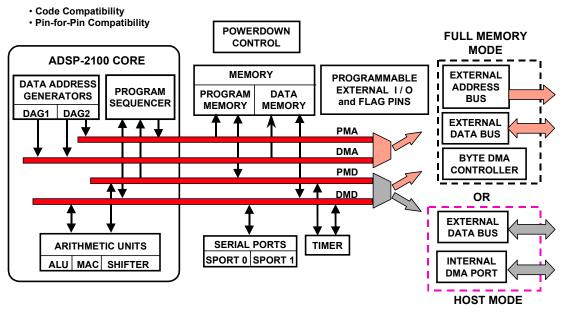
Supports Zero-Overhead Looping

```
ADSP21xx Example code:
CNTR = 10;
D0 endloop UNTIL CE;
I0(DACCONTROL) = AX0;
MR = MR + MX0 * MY0(SS), MX0 = DM(I0,M1), MY0 = PM(I4,M5);
endloop:
IF MV SET FL1;
IF EQ CALL mysubroutine;
```

Figure 7.13

ADSP-21xx-FAMILY ON-CHIP PERIPHERALS

The discussion so far has involved the core architecture of the fixed-point ADSP-21xx DSPs which is common to all members of the family. This section discusses the on-chip peripherals which have different configurations and options depending on the particular processor in the family. The ADSP-218x architecture is shown in Figure 7.14.



ADSP-218x FAMILY ARCHITECTURE

External bus features are multiplexed for 100-pin parts. All are available on 128-pin parts



ADSP-21xx ON-CHIP PERIPHERALS: MEMORY INTERFACE

- All Family Members use an Enhanced Harvard Architecture
 - Separate Program Memory and Data Memory Spaces
 - Can access Data Values in Program Memory
- Different Family Members Have Different Memory Configurations
- External Memory Interface Supports Both Fast and Slow Memories with Programmable Wait States
- Supports DSP Boot Loading from Byte Memory Space or from a Host Processor
- Supports Memory-Mapped Peripherals Through I / O Space
- Bus Request / Grant Mechanism for Shared External Bus

The 21xx-family comes with a variety of on-chip memory options, and the newer 218x family has up to 48K words of program memory and 56K words of data memory. All family members use the modified Harvard architecture which provides separate program and data memory and allows data to be stored in program memory. The external memory interface supports both fast and slow memories with programmable wait states. The ADSP-218x family also supports memory-mapped peripherals through I/O space.

All 21xx parts (except the ADSP-2105) have two double-buffered serial ports (SPORTs) for transmitting or receiving serial data. Each SPORT is bi-directional, full duplex, double-buffered, and has its own programmable serial clock. The SPORT word length can be configured from 3 to 16 bits. Data can be framed or unframed. Each SPORT generates an interrupt and supports A-law and u-law companding.

ADSP-21xx ON-CHIP PERIPHERALS: SERIAL PORTS (SPORTs)

- ADSP-21xx SPORTs Are Used For Synchronous Communication
- Full Duplex
- Fully Programmable
- Autobuffer/DMA Capability
- TDM Multi-Channel Capability
- A-law and u-law Companding
- Data Rates of 25 Mbits/sec and Above
- Glueless Interface to a Wide Range of Serial Peripherals or Processors
- **219x DSPs Add SPI and UART Serial Ports (With Boot Capability)**

Figure 7.16

The ADSP-218x-family internal direct-memory-access (IDMA) port supports booting from and runtime access by a host processor. This feature allows data to be transferred to and from internal memory in the background while continuing foreground processing. The IDMA port allows a host processor to access all of the DSPs internal memory without using mailbox registers. The IDMA port supports 16 and 24-bit words, and 24-bit transfers take two cycles to complete.

ADSP-21xx ON-CHIP PERIPHERALS: INTERNAL DMA (IDMA)

 Allows an External System to Access DSP Internal Memory
 External Device or DSP Can Specify Internal Starting Address
 Address Automatically Increments to Speed Throughput
 16-Bit Bus Supports Both Data and Instruction Transfers (219x has 8-Bit Bus Support)
 Single DSP Processor-Cycle Transfers
 Supports Power-On Booting

The ADSP-218x-family also has a byte memory interface which supports booting from and runtime access to 8-bit memories. It can access up to 4MB. This memory space takes the place of the boot memory space found on other ADSP-21xx family processors. Byte memory consists of 256 pages of 16K x 8 locations. This memory can be written and read in 24-bit, 16-bit, or 8-bit left or right justified transfers. Transfers happen in the background to the DSP internal memory by stealing cycles.

Figure 7.17

ADSP-21xx ON-CHIP PERIPHERALS: BYTE DMA PORT (BDMA)

- Provides Bulk Storage for Both Data and Program Code
- Can Access up to 4 Mbytes of External Code & Data
- Supports Multiple Data Formats
 - Automatic Data Packing/Unpacking to 16 and 24 bits
 - 8-Bit Transfers, Left- or Right-Justified
- Background Transfer to DSP Internal Memory
 - One Cycle per Word
 - DSP Specifies Destination/Source and Word Count
- Supports Power-On Booting
- Allows Multiple Code Segments
 - DSP Can Overlay Code Sections
 - Processor Can Run During Transfer, or Halt and Restart

Figure 7.18

The ADSP-217x, ADSP-218x, and ADSP-21msp5x devices provide a powerdown feature that allows the processor to enter a very low power state (less than 1mW) through hardware or software control. This feature is extremely useful for battery-powered applications. During some of the powerdown modes, the internal clocks are disabled, but the processor registers and memory are maintained.

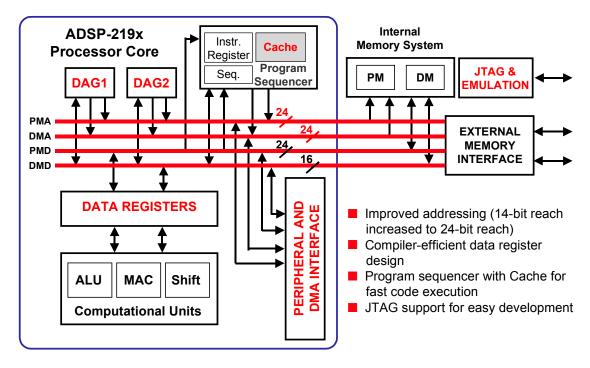
ADSP-21xx INTERNAL PERIPHERALS: POWERDOWN

- Non-Maskable Interrupt
 - + Hardware Pin (PWD), or Software Forced
- Holds Processor in CMOS Standby
- Rapid CLKIN-Cycle Recovery
- Acknowledge Handshake (PWDACK)
- Ideal for Battery-Powered Applications
- 219x is Fully Static

Figure 7.19

From the above discussions it should be obvious that the ADI DSPs are designed for maximum efficiency when performing typical DSP functions such as FFTs or digital filtering. ADI DSPs can perform several operations in an instruction cycle as has been shown in the above filter example. DSPs are often rated in terms of Millions of Instructions per Second, or MIPS. However, the MIPS rating does not tell the entire story. For example if processor A has a instruction rate of 50 MIPS and can perform 1 operation per instruction, it can perform 50 Million Operations per Second, or 50 MOPS. Now assume that processor B has an instruction rate of 20 MIPS, but can perform 4 operations per instruction. Processor B can therefore perform 80 Million Operations per Second, or 80 MOPS, and is actually more efficient than processor A. A better way to evaluate DSP performance is to use well defined benchmarks such an FIR filter with a prescribed number of taps or an FFT of known size. Benchmark comparisons eliminate the confusion often associated with MIPS or MOPS ratings alone and are discussed later in this section. Even this form of benchmarking does not give a true comparison of performance between two processors. Analysis of the target system requirements, processor architecture, memory needs and other factors must be considered.

The ADSP-219x family maintains code compatibility with the ADSP-218x. Streamlined for faster processing and improved C-compiler efficiency, this new family will include DSPs with speeds between 100 and 300MIPS and power consumption as low as 0.4mA/MIP. JTAG support is also included to provide a more robust software emulation and test capability. A block diagram of the family is shown in Figure 7.20.



ADSP-219x SERIES ARCHITECTURE

Figure 7.20

The address reach of the ADSP-219x series has been extended from the ADSP-218x's 14-bit reach to a 24-bit reach. This supports 64K word direct memory addressing or 16M word paged memory addressing. All existing addressing modes are supported, and five new DAG addressing modes have been added.

Many of the enhancements in the ADSP-219x are designed to improve compiler efficiency. A global register allocator and support for register file-like operand access reduces spills and reduces reliance on the local stack. The compiler features DSP intrinsic support including fractional and complex. On-chip cache memory has also been added.

The ADSP-219x core will serve as a key DSP technology for ADI's 16-bit general purpose DSP offerings, and embedded DSP solutions, where application-specific circuitry and software are custom-designed to a customer's precise requirements. For performance-driven applications, multiple cores will be integrated together on a single die. In the future, four cores will be combined in a family of devices capable of delivering 1.2 billion MACs per second per square inch. Power-conscious designers will appreciate operating currents of 0.15mA/MIPS.

ADSP-219x FAMILY KEY SPECIFICATIONS

- Code Compatible
 - Compatible with ADSP-218x Series
 - Single Cycle Instruction Execution, Zero-Over Head Looping, Single Cycle Context Switch
- Performance
 - Architectural Performance Beyond 300 MIPs.
 - Fully Transparent Instruction Cache
- Compiler-Friendly and Tool-Friendly
 - 64K Word Direct and 16 Mword Paged Memory Support
 - 5 New DAG Addressing Modes
 - Register File-Like Operand Access
 - JTAG Interface Support

Figure 7.21

The history of the Analog Devices 16-bit fixed-point DSP family is shown in Figure 7.22. Notice the migration in performance as well as improvements in power dissipation and packaging while maintaining code compatibility between the various devices. The newer families offer 3.3V (L-series) and 2.5V (M-series) operation for further efficiency. Earlier DSPs were packaged in expensive pin-grid-array (PGA) packages or plastic-leaded-chip-carriers (PLCCs), but these have been largely replaced by plastic-quad-flat-packs (PQFPs), and more recently by thin (1.6mm) quad-flat-packs (TQFPs). Note: In 1998, JEDEC changed the specifications for the TQFP package designation, assigning it to packages 1.0mm thick. *Previously labeled TQFP packages (1.6mm thick) are now designated as LQFP per the JEDEC specification*.

The 144-ball miniBGA package (see Figure 7.23) from ADI represents innovative packaging combined with low power consumption (0.4mA per MIP), and allows 75MIPS operation and more than 2M bits of SRAM in a 1cm² package which is 1.35mm thick. For example, the 75MIPS ADSP-2188M has 48K of 24-bit program memory and 56K of 16-bit data memory for a total of $48K \times 24 + 56K \times 16 = 2028K$ bits with a power dissipation of less than 100mW.

16-BIT DSP FAMILY TREE: HISTORY OF IMPROVEMENTS IN <u>PACKAGING</u> - <u>POWER</u> - <u>PERFORMANCE</u>

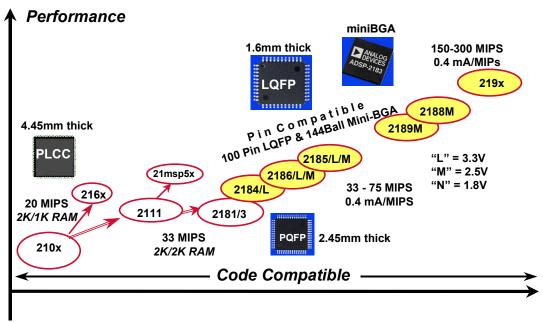


Figure 7.22

'M' SERIES OFFERS THE LARGEST MIPS/MEMORY DENSITY WITH miniBGA PACKAGING!!!

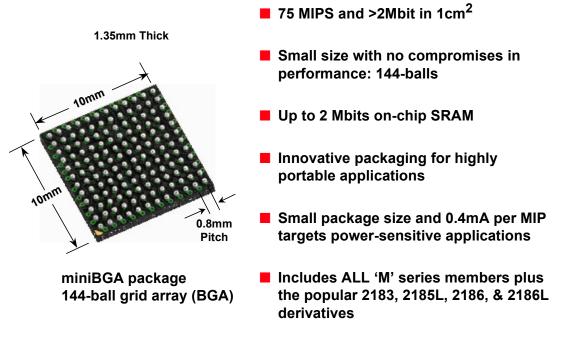


Figure 7.23

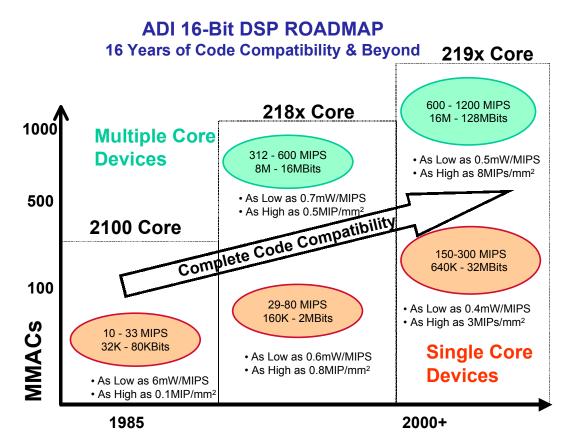


Figure 7.24

ADSP-218x ROADMAP

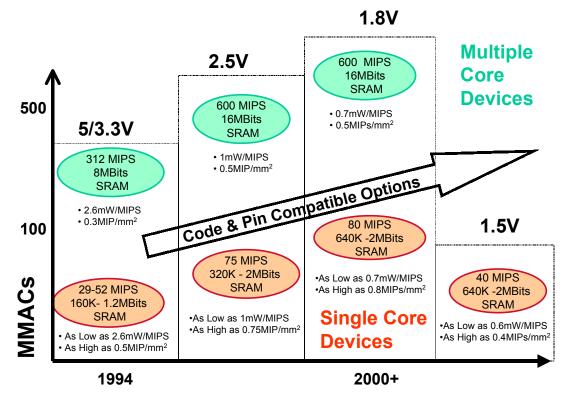
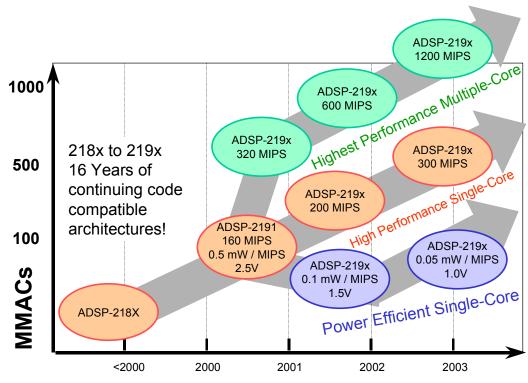


Figure 7.25



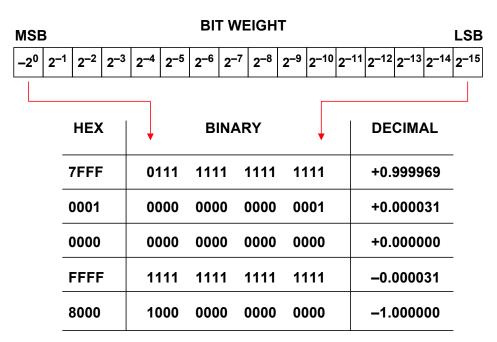
ADSP-219x ROADMAP

Figure 7.26

FIXED-POINT VERSUS FLOATING POINT

DSP arithmetic can be divided into two catagories: *fixed-point* and *floating-point*. These refer to the format used to store and manipulate the numbers within the devices. The Analog Devices' fixed-point DSPs such as those discussed so far represent each number with 16 bits. There are four common ways that $2^{16} = 65,536$ possible bit patterns can represent a number. In *unsigned integer format*, the stored number takes on any integer value from 0 to 65,536. In *signed integer format*, two's complement is used to make the range include negative numbers, from -32,768 to +32,767. Using *unsigned fractional format*, the 65,536 levels are spread uniformly between 0 and +1. Finally, *signed fractional format* allows negative numbers, with 65,536 levels equally spaced between -1 and +1.

The ADSP-21xx family arithmetic is optimized for the signed fractional format denoted by 1.15 ("one dot fifteen"). In the 1.15 format, there is one sign bit (the MSB) and fifteen fractional bits representing values from -1 up to 1 LSB less than +1 as shown in Figure 7.27.



16-BIT FIXED POINT ARITHMETIC FRACTIONAL 1.15 FORMAT

Figure 7.27

This convention can be generalized as "I.Q", in which I is the number of bits to the left of the radix point, and Q is the number of bits to the right. For example, full unsigned integer number representation is 16.0 format. For most signal processing applications, however, fractional numbers (1.15) are assumed. Fractional numbers have the advantage that the product of two fractional numbers is smaller than either of the numbers.

By comparison, floating point DSPs typically use a minimum of 32 bits to represent each number. This results in many more possible numbers than for 16-bit fixed point, $2^{32} = 4,294,967,296$ to be exact. More importantly, floating point greatly increases the range of values that can be expressed. The most common floating point standard is ANSI/IEEE Standard 754-1985, where the largest and smallest numbers allowed by the standard are $\pm 3.4 \times 10^{38}$ and $\pm 1.2 \times 10^{-38}$, respectively. Note that the 754 standard reserves some of the possible range to free up bit patterns which allow other special classes of numbers such as ± 0 and $\pm \infty$, for example.

The IEEE-754 floating point standard is described in more detail in Figure 7.28. The 32-bit word is divided into a sign bit, S, an 8-bit exponent, E, and a 23-bit mantissa, M. The relationship between the decimal and binary IEEE-754 floating point equivalent is given by the equation:

NUMBER₁₀ =
$$(-1)^{S} \times 1.M \times 2^{(E-127)}$$
.

Notice that the "1." is assumed to precede the "M", and that a "bias" of 127 is subtracted from the exponent "E" so that "E" is always a positive number.

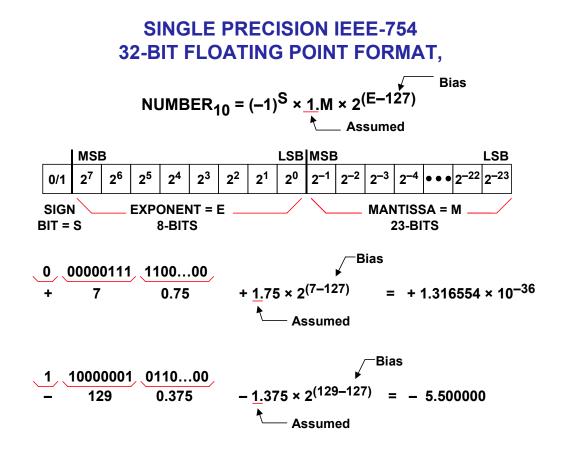


Figure 7.28

In the case of *extended precision* floating point arithmetic, there is one sign bit, the mantissa is 31-bits, the exponent is 11-bits, and the total word length is 43-bits.

Extended precision thus adds 8 bits of dynamic range to the mantissa and can execute almost as fast as single precision, since accumulators can readily be extended beyond 32 bits. On the other hand, true 64-bit *double precision* (52-bit mantissa, 11-bit exponent, and 1 sign bit) requires extra processor cycles. Requirements for double precision are rare in most DSP applications.

Many DSP applications can benefit from the extra dynamic range provided by 32-bit floating point arithmetic. In addition, programming floating point processors is generally easier, because fixed-point problems such as overflow, underflow, data scaling, and round-off error are minimized, if not completely eliminated. Although the floating point DSP may cost slightly more than the fixed point DSP, development time may well be shorter with floating point.

While all floating point DSPs can also handle fixed point numbers (required to implement counters, loops, and ADC/DAC signals), this doesn't necessarily mean that fixed point math will be carried out as quickly as the floating point operations; it depends on the internal DSP architectures. For instance, the SHARC DSPs from Analog Devices are optimized for both floating point and fixed point operations, and execute them with equal efficiency. For this reason, the SHARC devices are often referred to as "32-bit DSPs" rather than just "Floating Point."

FIXED POINT VS. FLOATING POINT ARITHMETIC

- 16-Bit Fixed-Point:
 - ◆ 2¹⁶ = 65,536 Possible Numbers
- **32-Bit Floating Point:**
 - Biggest Number: ±6.8 × 10³⁸ 754 Std: ±3.4 × 10³⁸
 - ◆ Smallest Number: ±5.9 × 10⁻³⁹ 754 Std: ±1.2 × 10⁻³⁸
- Extended-Precision (40-Bits: Sign + 8-Bit Exponent + 31-Bit Mantissa)
- Double-Precision (64-Bits: Sign + 11-Bit Exponent + 52-Bit Mantissa)
- 32-Bit Floating Point
 - More Precision
 - Much Larger Dynamic Range
 - Easier to Program

Figure 7.29

ADI SHARC® FLOATING POINT DSPS

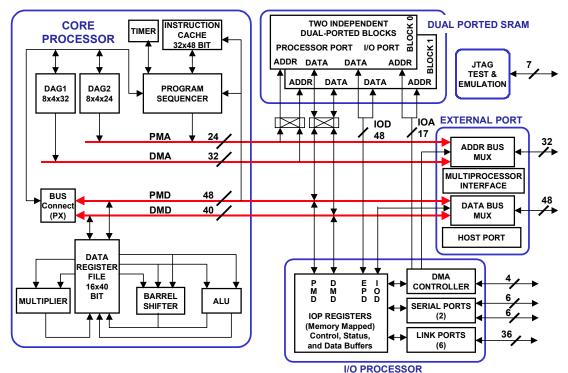
The ADSP-2106x Super Harvard Architecture SHARC is a high performance 32-bit DSP. The SHARC builds on the ADSP-21000 family DSP core to form a complete system-on-a-chip, adding a dual-ported on-chip SRAM and integrated I/O peripherals supported by a dedicated I/O bus. With its on-chip instruction cache, the processor can execute every instruction in a single cycle. Four independent buses for dual data, instructions, and I/O plus crossbar switch memory connections comprise the Super Harvard Architecture of the ADSP-2106x shown in Figure 7.30.

A general purpose data register file is used for transferring data between the computation units and the data buses, and for storing intermediate results. The register file has two sets (primary and alternate) of 16 registers each, for fast context switching. All of the registers are 40 bits wide. The register file, combined with the core processor's super Harvard architecture, allows unconstrained data flow between computation units and internal memory.

The ADSP-2106x SHARC processors address the five central requirements for DSPs established in the ADSP-21xx family of 16-bit fixed point DSPs: (1) fast, flexible arithmetic computation units, (2) unconstrained data flow to and from the computation units, (3) extended precision and dynamic range in the computation units, (4) dual address generators, and (5) efficient program sequencing with zero-overhead looping.

The program sequencer includes a 32-word instruction cache that enables three-bus operation for fetching an instruction and two data values. The cache is selective – only instructions whose fetches conflict with program memory data accesses are cached. This allows full-speed multiply-accumulates and FFT butterfly processing.

ADI SUPER HARVARD ARCHITECTURE (SHARC®) 32-BIT DSP ARCHITECTURE FOR ADSP-2106x FAMILY



NOT ROOLD

Figure 7.30

SHARC KEY FEATURES

- 100MHz Core / 300 MFLOPS Peak
- Parallel Operation of: Multiplier, ALU, 2 Address Generators & Sequencer
 - No Arithmetic Pipeline; All Computations Are Single-Cycle
- High Precision and Extended Dynamic Range
 - ♦ 32/40-Bit IEEE Floating-Point Math
 - 32-Bit Fixed-Point MAC's with 64-Bit Product & 80-Bit Accumulation
- Single-Cycle Transfers with Dual-Ported Memory Structures
 - Supported by Cache Memory and Enhanced Harvard Architecture
- Glueless Multiprocessing Features
- **JTAG Test and Emulation Port**
- DMA Controller, Serial Ports, Link Ports, External Bus, SDRAM Controller, Timers

SHARC® THE LEADER IN FLOATING POINT DSP

- SHARC is the de facto standard in multiprocessing
- ADSP-21160 continues SHARC leadership in multiprocessing
- ADSP-21065L is the right choice for low-cost floating point

SUPER HARVARD ARCHITECTURE:
Balancing Memory, I/O and Computational Power...
High Performance Computation Unit
Four Bus Performance
Fetch Next Instruction
Access 2 data values

- Perform DMA for I/O
- Memory Architecture
- Non Intrusive DMA



Figure 7.32

The ADSP-2106x family execute all instructions in a single cycle. They handle 32-bit IEEE floating point format, 32-bit integer and fractional fixed point formats (twoscomplement and unsigned), and extended precision 40-bit IEEE floating point format. The processors carry extended precision throughout their computation units, minimizing intermediate data truncation errors. When working with data onchip, the extended precision 32-bit mantissa can be transferred to and from all computation units. The 40-bit data bus may be extended off-chip if desired. The fixed point formats have an 80-bit accumulator for true 32-bit fixed point computations.

The ADSP-2106x has a super Harvard architecture combined with a 10-port data register file. In every cycle, (1) two operands can be read or written to or from the register file, (2) two operands can be supplied to the ALU, (3) two operands can be supplied to the multiplier, and (4) two results can be received from the ALU and multiplier.

The ADSP-2106x family instruction set provides a wide variety of programming capabilities. Multifunction instructions enable computations in parallel with data transfers, as well as simultaneous multiplier and ALU operations.

The ADSP-21060 contains 4 Mbits of on-chip SRAM, organized as two blocks of 2 Mbits each, which can be configured for different combinations of code and data storage. The ADSP-21062, ADSP-21061 and ADSP-21065 each include 2 Mbit, 1Mbit and 544Kbits of on-chip SRAM, respectively. Each memory block is dual-ported for single-cycle, independent accesses by the core processor and I/O processor or DMA controller. The dual-ported memory and separate on-chip buses allow two data transfers from the core and one from I/O, all in a single cycle.

While each memory block can store combinations of code and data, accesses are most efficient when one block stores instructions and data, using the DM bus for transfers, and the other block stores instructions and data, using the PM bus for transfers. Using the DM bus and PM bus in this way, with one dedicated to each memory block, assures single-cycle execution with two data transfers. In this case, the instruction must be available in the cache. Single-cycle execution is also maintained when one of the data operands is transferred to or from off-chip, via the ADSP-2106x's external port.

The ADSP-2106x's external port provides the processor's interface to off-chip memory and peripherals. The 4 Gword off-chip address space is included in the ADSP-2106x's unified address space. The separate on-chip buses – for PM addresses, PM data, and DM addresses, DM data, I/O addresses, and I/O data – are multiplexed at the external port to create an external system bus with a single 32-bit address bus and a single 48-bit data bus. The ADSP-2106x provides programmable memory wait states and external memory acknowledge controls to allow interfacing to DRAM and peripherals with variable access, hold, and disable time requirements.

The ADSP-2106x's host interface allows easy connection to standard microprocessor buses, both 16-bit and 32-bit, with little additional hardware required. Four channels of DMA are available for the host interface; code and data transfers are accomplished with low software overhead. The host can directly read and write the internal memory of the ADSP-2106x, and can access the DMA channel setup and mailbox registers. Vector interrupt support is provided for efficient execution of host commands.

The ADS-2106x offers powerful features tailored to multiprocessing DSP systems. The unified address space allows direct interprocessor accesses of each ADSP-2106x's internal memory. Distributed bus arbitration logic is included on-chip for simple, glueless connection of systems containing up to six ADSP-2106xs and a host processor. Master processor changeover incurs only one cycle of overhead. Bus arbitration is selectable as either fixed or rotating priority. Maximum throughput for interprocessor data transfer is 240 Mbytes/second (with a 40MHz clock) over the link ports or external port.

The ADSP-2106x's I/O Processor (IOP) includes two serial ports, six 4-bit link ports, and a DMA controller. The ADSP-2106x features two synchronous serial ports that provide an inexpensive interface to a wide variety of digital and mixed-signal peripheral devices. The serial ports can operate at the full external clock rate of the processor, providing each with a maximum data rate of 50 Mbit/second. Independent transmit and receive functions provide greater flexibility for serial communications. Serial port data can be automatically transferred to and from onchip memory via DMA. Each of the serial ports offers a TDM multichannel mode. They offer optional μ -law or A-law companding. Serial port clocks and frame syncs can be internally or externally generated.

The ADSP-21060 and ADSP-21062 feature six 4-bit link ports that provide additional I/O capabilities. The link ports can be clocked twice per cycle, allowing each to transfer 8 bits per cycle. Link port I/O is especially useful for point-to-point interprocessor communication in multiprocessing systems. The link ports can operate independently and simultaneously, with a maximum data throughput of 240 Mbytes/second. Link port data is packed into 32-bit or 48-bit words, and can be directly read by the core processor or DMA-transferred to on-chip memory. Each link port has its own double-buffered input and output registers. Clock/acknowledge handshaking controls link port transfers. Transfers are programmable as either transmit or receive. There are no link ports on the ADSP-21061 or ADSP-21065 devices.

The ADSP-2106x's on-chip DMA controller allows zero-overhead data transfers without processor intervention. The DMA controller operates independently and invisibly to the processor core, allowing DMA operations to occur while the core is simultaneously executing its program. Both code and data can be downloaded to the ADSP-2106x using DMA transfers. DMA transfers can occur between the ADSP-2106x's internal memory and external memory, external peripherals, or a host processor. DMA transfers can also occur between the ADSP-2106x's internal memory and its serial ports or link ports. DMA transfers between external memory and external memory and external memory and external memory and its serial ports or link ports. DMA transfers between external memory and external memory and external memory and its serial ports or link ports. DMA transfers between external memory and its serial ports or link ports. DMA transfers between external memory and external memory

The internal memory of the ADSP-2106x can be booted at system powerup from an 8-bit EPROM or a host processor. Additionally, the ADSP-21060 and the ADSP-21062 can also be booted through one of the link ports. Both 32-bit and 16-bit host processors can be used for booting.

The ADSP-2106x supports the IEEE standard P1149.1 Joint Test Action Group (JTAG) standard for system test. This standard defines a method for serially scanning the I/O status of each component in a system. In-circuit emulators also use the JTAG serial port to access the processor's on-chip emulation features. EZ-ICE Emulators use the JTAG test access port to monitor and control the target board processor during emulation. The EZ-ICE in-circuit emulator provides full-speed emulation to enable inspection and modification of memory, registers, and processor stacks. Use of the processor's JTAG interface assures non-intrusive in-circuit emulation – the emulator does not affect target system loading or timing.

The SHARC architecture avoids processor bottlenecks by balancing core, memory, I/O processor, and peripherals as shown in Figure 7.30. The core supports 32-bit fixed and floating point data. The memory contributes to the balance by offering large size and dual ports. The core can access data from one port, and the other port is used to move data to and from the I/O processor. The I/O processor moves data to and from the peripherals to the internal memory using zero overhead DMAs. These operations run simultaneously to the core operation.

ADSP-2116X SINGLE-INSTRUCTION, MULTIPLE-DATA (SIMD) CORE ARCHITECTURE

The ADSP-21160 is the first member of the second-generation ADI 32-bit DSPs. Its core architecture is shown in Figure 7.33. Notice that the core is similar to the ADSP-2106x core except for the width of the buses and the addition of a second computational unit complete with its own multiplier, ALU, shifter, and register file. This architecture is called *single-instruction, multiple-data* (SIMD) as opposed to *single-instruction, single-data* (SISD). The second computational unit allows the DSP to process multiple data streams in parallel. The core operates at up to 100 MIPS. At 100MHz clock operation the core is capable of 400 MFLOPS (millions of floating point operations per second) sustained and 600 MFLOPS peak operation. SIMD is a natural next step in increasing performance for ADI DSPs. Because their basic architecture already allows single instruction and multiple data. The SIMD architectural extension allows code-compatible higher performance parts.

ADSP-2116x CORE PROCESSOR FEATURING SINGLE-INSTRUCTION, MULTIPLE-DATA (SIMD)

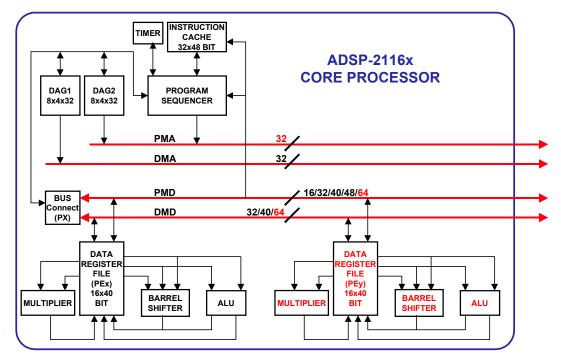


Figure 7.33

DSP HARDWARE

The SIMD features of the ADSP-2116x include two computational units (PEx, PEy) and double data-word size buses (DMD and PMD). The primary processing element, PEx, is always enabled. The secondary processing element, PEy, is mode control enabled. The double wide data buses provide each computational unit with its own data set in each cycle. With SIMD enabled, both processing elements execute the same instruction each cycle (that's the Single-Instruction), but they execute that instruction using different data (that's the Multiple-Data). The SIMD-based performance increase appears in algorithms that can be optimized by splitting the processing of data between the two computational units. By taking advantage of the second computational unit the cycle time can be cut in half compared to the SISD approach for many algorithms.

The ADSP-21160 has a complete set of integrated peripherals: I/O Processor, 4 Mbyte on-chip dual-ported SRAM, glueless multiprocessing features, and ports (serial, link, external bus, host, and JTAG). Power dissipation is approximately 2W at 100MHz in a 400-ball 27 × 27mm PBGA package. The complete SHARC family roadmap is shown in Figure 7.35.

ADSP-21160 32-BIT SHARC KEY FEATURES

- SIMD (Single-Instruction, Multiple-Data) Architecture
- Code Compatible with ADSP-2106x Family
- 100 MHz Core / 600 MFLOPS Peak
- On-Chip Peripherals Similar to ADSP-2106x Family
- Dual-Ported 4Mbit SRAM
- Glueless Multiprocessing Features
- 400-Ball, PBGA 27×27mm Package

Figure 7.34

Figure 7.36 shows some typical coding using the SHARC family of DSPs. Note the algebraic syntax of the assembly language which facilitates coding of algorithms and reading the code after it is written. In a single cycle, the SHARC performs multiplication, addition, subtraction, memory read, memory write, and address pointer updates. In the same cycle, the I/O processor can transfer data to and from the serial port, the link ports, memory or DMA, and update the DMA pointer.

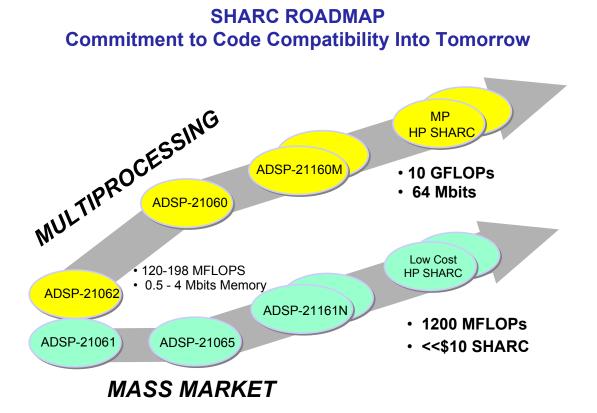


Figure 7.35

EXAMPLE: SHARC MULTI-FUNCTION INSTRUCTION

- f11 = f1 * f7, f3 = f9 + f14, f9 = f9 f14, dm (i2, m0) = f13, f7 = pm (i8, m8);
- In this Single-Cycle Instruction the SHARC Performs:
 - 1 (2) Multiply
 - 1 (2) Addition
 - 1 (2) Subtraction
- () = ADSP-2116x SIMD DSP
- 1 (2) Memory Read
- 1 (2) Memory Write
- 2 Address Pointer Updates
- Plus the I/O Processor Performs:
 - Active Serial Port Channels: Transmit and Receive on all Ports
 - 6 Active Link Ports if Present
 - Memory DMA
 - 2 (4) DMA Pointer Updates
- The Algebraic Syntax of the Assembly Language Facilitates Coding of DSP Algorithms

MULTIPROCESSING USING SHARCS

Analog Devices' SHARC DSPs such as the ADSP-21160 are optimized for multiprocessing applications such as telephony, medical imaging, radar/sonar, communications, 3D graphics and imaging. Figure 7.37 shows SHARC benchmark performance on common DSP algorithms.

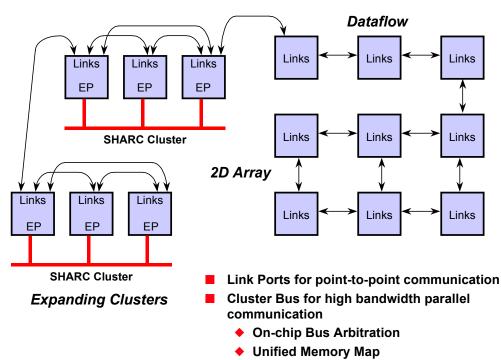
	ADSP-21065L	ADSP-21160	ADSP-21160 SIMD/
	SHARC	SISD	multiple channels
Clock Cycle	66 MHz	100 MHz	100 MHz
Instruction Cycle Time	15 ns	10 ns	10 ns
MFLOPS Sustained	132 MFLOPS	200 MFLOPS	400 MFLOPS
MFLOPS Peak	198 MFLOPS	300 MFLOPS	600 MFLOPS
1024 Point Complex FFT	274 µs	180 µs	90 µs
(Radix 4, with reversal)			
FIR Filter (per tap)	15 ns	10 ns	5 ns
IIR Filter (per biquad)	60 ns	40 ns	20 ns
Matrix Multiply			
(pipelined)			
[3x3] * [3x1]	135 ns	90 ns	45 ns
[4x4] * [4x1]	240 ns	160 ns	80 ns
Divide (y/x)	90 ns	60 ns	30 ns
Square Root	135 ns	90 ns	45 ns

DSP BENCHMARKS FOR SHARC FAMILY

Figure 7.37

Multiprocessor systems typically use one or both of two methods to communicate between processor nodes. One method uses dedicated point-to-point communication channels. This method is often called *data-flow multiprocessing*. In the other method, nodes communicate through a single shared global memory via a parallel bus. The SHARC family supports the implementation of point-to-point communication through its six link ports called *link port multiprocessing*. It also supports an enhanced version of shared parallel bus communication called *cluster multiprocessing*.

For applications that require high computational bandwidth, but only limited flexibility, data flow multiprocessing is the best solution. The DSP algorithm is partitioned sequentially across several processors and data is passed directly across them as shown on the right side of Figure 7.38. The SHARC is ideally suited for data flow multiprocessing applications because it elminates the need for interprocessor data FIFOs and external memory. Each SHARC has six link ports allowing 2D and 3D arrays as well as traditional data flow. The internal memory of the SHARC is usually large enough to contain both code and data for most applications using this topology. All a data flow system requires are a number of SHARC processors and point-to-point signals connecting them.



MULTIPROCESSOR COMMUNICATION EXAMPLES FOR SHARC

Figure 7.38

EXTERNAL PORT VERSUS LINK PORT COMMUNICATIONS

Advantages of External Ports (EP)

- Communications through the EP has the Highest Bandwidth between two SHARCs (400 Mbytes/s)
- Allows up to 6 SHARCs and a Host to share the EP
- The EP offers flexible communication of data and control information
- The Shared Memory model allows a simple software structure
- Advantage of Link Ports
 - Each Link Port provides independent, 100 MBytes/s communication between two SHARCs
 - Up to 6 Link Ports (600 MBytes/s)
 - Easily scalable to any number of SHARCs
- Both Link Port and EP communications can be used simultaneously

Cluster multiprocessing is best suited for applications where a fair amount of flexibility is required. This is especially true when a system must be able to support a variety of different tasks, some of which may be running concurrently. SHARC processors also have an on-chip host interface that allows a cluster to be easily interfaced to a host processor or even to another cluster.

Cluster multiprocessing systems include multiple SHARC processors connected by a parallel bus that allows interprocessor access of on-chip memory as well as access to shared global memory. In a typical cluster of SHARCs, up to six ADSP-21160 processors and a host can arbitrate for the bus. The on-chip bus arbitration logic allows these processors to share the common bus. The SHARC's other on-chip features help eliminate the need for any extra glue hardware in the cluster multiprocessor configuration. External memory, both local and global, can frequently be eliminated in this type of system.

TIGERSHARCTM: THE ADSP-TS001 STATIC SUPERSCALAR DSP

The ADSP-TS001 is the first DSP from Analog Devices to use the new TigerSHARC[™] static superscalar architecture. The TigerSHARC targets telecommunications infrastructure equipment with a new level of integration and the unique ability to process 8-, 16-, 32-bit fixed and floating-point data types on a single chip. Each of these data types is critical to the next generation of telecommunications protocols currently under development, including IMT-2000 (also known as 3G wireless) and xDSL (digital subscriber line). Unlike any other DSP, the ADSP-TS001 has the unique ability to accelerate processing speed based on the data type. Moreover, the chip delivers the highest performance floating-point processing.

In telecommunications infrastructure equipment, voice coder and channel coder protocols are developed around 16-bit data types. To improve signal quality, many telecom applications employ line equalization and echo cancellation techniques that boost overall signal quality and system performance. These algorithms benefit from the added precision of 32-bit and floating-point data processing. The 8-bit native support is well suited to the commonly used Viterbi channel decoder algorithm, as well as image processing where it is more straightforward and cost-effective to represent red, green, and blue components of the signal with 8-bit data types. Many of these applications require high levels of performance and may require algorithms to be executed consecutively or even concurrently. The end application determines the exact requirements. The flexibility of the TigerSHARC architecture enables the software engineer to match the application precision requirements without any loss of system performance. In the TigerSHARC, performance is traded directly against numerical precision.

The TigerSHARC architecture uses key elements from multiple microprocessor types – RISC (reduced instruction set computer), VLIW (very long instruction word), and DSP in order to provide the highest performance digital signal processing engine. The new architecture leverages existing DSP product attributes such as fast and deterministic execution cycles, highly responsive interrupts, and an excellent peripheral interface to support large core computation rates and a high data rate I/O. To achieve excellent core performance, RISC-like features such as load/store operations, a deeply pipelined sequencer with branch prediction, and large interlocked register files are introduced. Additionally, the VLIW (very long instruction word) attributes offer more efficient use of code space, especially for control code.

TigerSHARC®: ANALOG DEVICES' NEW STATIC SUPERSCALER DSP ARCHITECTURE

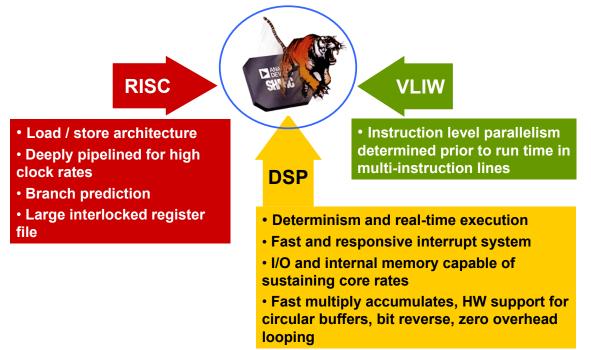


Figure 7.40

TigerSHARC KEY ARCHITECTURAL FEATURES

Core

- 1200 MMACs/s @ 150 MHz -- 16-Bit Fixed Point
- **300 MMACs/s @150 MHz -- 32-Bit Floating Point**
- 900 MFLOPS -- 32-Bit Floating Point

Memory

6 Mbits of on-chip SRAM organized in a unified memory map as opposed to the traditional Harvard architecture.

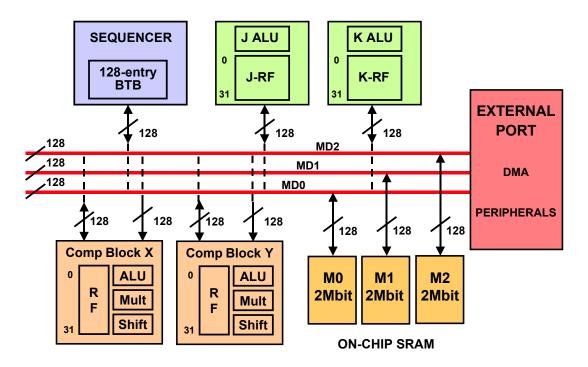
I/O, Peripherals, & Package

- **600** Mbytes/s transfer rate through external bus.
- 600 Mbytes/s aggregate transfer rate through 4 Link Ports
- Glueless multiprocessor cluster support for up to 8 ADSP-TS001s
- 4 General Purpose I/O Ports
- SDRAM Controller
- 360 Ball, SBGA Package 35×35mm

Finally, to supply all functional blocks with instructions, clever management of the instruction word is necessary. Specifically, multiple instructions must be dispatched to processing units simultaneously, and functional parallelism must be calculated prior to runtime.

By incorporating the best of all worlds, the TigerSHARC architecture will provide a state of the art platform for the most demanding signal processing applications.

The TigerSHARC core shown in Figure 7.42 consists of multiple functional blocks: computation blocks, memory, integer ALUs, and a sequencer. There are two computational blocks (X and Y) in the TigerSHARC architecture, each containing a multiplier, ALU, and 64-bit shifter. With the resources in these blocks, it is possible to execute eight 40-bit MACs on 16-bit data, two 40-bit MACs on 16-bit complex data, or two 80-bit MACs on 32-bit data – all in a single cycle. TigerSHARC is a register-based load/store architecture, where each computational block has access to a fully orthogonal 32-word register file.



ADSP-TS001 TigerSHARC® ARCHITECTURE

Figure 7.42

The TigerSHARC DSP features a short-vector memory architecture organized in three 128-bit wide banks. Quad, long, and normal word accesses move data from the memory banks to the *register* files for operations. In a given cycle, four 32-bit instruction words can be fetched, and 256 bits of data can be loaded to the register files or stored into memory. Data in 8-, 16-, or 32-bit words can be stored in contiguous, packed memory. Internal and external memories are organized in a unified memory map which leaves specific partitioning to the programmer. The internal memory bandwidth for data and instructions is 7.2 Gbytes/second when operating on a 150MHz clock.

Two integer ALUs are available for data addressing and pointer updates. They support circular buffering and bit reversal, and each has its own 32-word register file. More than simple data address generations units, both integer ALUs support general purpose integer computations. The general purpose nature of the integer ALUs benefits the compiler efficiency and increases programming flexibility.

The TigerSHARC architecture is designated *static superscalar*, as it executes up to four 32-bit instructions per clock cycle, and the programmer has the flexibility to issue individual instructions to each of the computational units. The sequencer supports predicted execution, where any individual instruction executes according to the result of a previously defined condition. The same instruction can be executed by the two computation blocks concurrently using different data values (this is called SIMD – single-instruction multiple-data operation).

The TigerSHARC architecture enables native operation using 8-, 16-, or 32-bit data values. The overall processor performance increases as the level of data precision decreases.

The inclusion of a *branch target buffer* (BTB) and *static branch prediction* logic eliminates the programming task of filling the instruction pipeline after branch instructions. If seen before, the branch is taken in a single cycle.

Three internal 128-bit wide busses ensure a large data bandwidth between internal functional blocks and external peripherals. The three-bus structure matches typical mathematical instructions which require two inputs and compute one output. The programming model is orthogonal and provides for deterministic interrupts.

The TigerSHARC architecture is free of hardware modes. This eliminates wasted cycles and simplifies compiler operation. The instruction set directly supports all DSP, image, and video processing arithmetic types including signed, unsigned, fractional, and integer data types. There is optional saturation (clipping) arithmetic for all cases.

At 150MHz, the ADSP-TS001 offers the highest integer and floating point performance of any SHARC product. Additionally, at 6 Mbits of on-chip SRAM, Analog Devices has increased its level of memory integration by 50% over previous SHARC family members. The migration to smaller process geometries will enable ADI to increase clock frequencies and integrate additional memory for future product derivatives.

TigerSHARC KEY FEATURES

- Execution of 1 to 4 32-Bit Instructions Per Clock Cycle
- Single-Instruction Multiple Data (SIMD) Operations Supported by Two Computation Blocks
- Multiple Data Type Computation Blocks
 - Each With Register File, MAC, ALU, Shifter
 - 32/40-Bit Floating or 32-Bit Fixed Point Operations (6 Per Cycle)
 - 16-Bit (24 Per Cycle) or 8-Bit (32 Per Cycle) Operations
- Static Branch Prediction Mechanism, with 128-Entry Branch Target Buffer (BTB)
- Internal Bandwidth of 7.2 Gbytes/second
- Simple and Fully Interruptible Programming Model

Figure 7.43

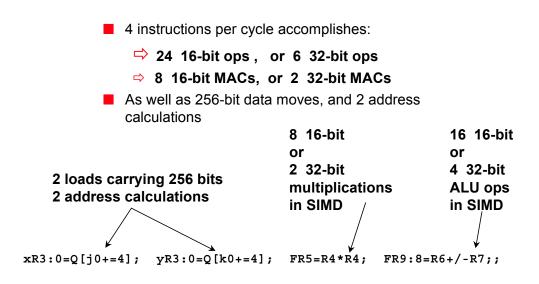
The ADSP-TS001 reduces total material costs by integrating multiple I/O and peripheral functions that reduce or eliminate the need for external glue logic and support chips. Specifically, the ADSP-TS001 at 150MHz integrates four glueless link ports with an aggregate transfer rate of 600 Mbytes/s, glueless multiprocessor cluster interface support for up to 8 ADSP-TS001s, an SDRAM controller, and a JTAG controller. This unprecedented functionality is packaged in a 35 by 35mm 360 ball SBGA package.

Typical computation rates and coding details of the TigerSHARC are shown in Figure 7.44. Four 32-bit instructions are executed in parallel forming one 128-bit instruction line. The entire instruction line is executed in one cycle. This example assembly code is for a single line and is performing the following:

xR3:0=Q[j0+=4];//	load 4 registers (xR0,xR1,xR2,xR3) in the X register file from memory
yR3:0=Q[k0+=4];//	load 4 registers in the Y register file from memory
FR5=R4*R4; //	multiply 2 32-bit floats in X computational block and 2 more in Y (2 multiplies)
FR9:8=R6+/-R7;;//	add and subtract in both X and Y computational blocks (4 ALU operations)

A single semicolon separates each 32-bit instruction, and a double semicolon indicates the end of an instruction line. This particular example shows the syntax for 32-bit floating point multiplies and ALU operations. Parallel 16-bit operands can easily be specified by using the "S" for "short" prefix instead of the "F" for the "float" prefix. J0 and K0 are IALU registers being used as indirect address pointers for the memory reads.

TigerSHARC PEAK COMPUTATION RATES





DSP programmers demand and require the capability to program in both high level languages and low level assembly language. The determination of programming language is dependent upon a number of factors including speed performance, memory size, and time to market considerations. Ultimately, however, the whole DSP product should incorporate features that enable user friendly coding in both high level and low level languages. The TigerSHARC architecture does indeed meet these requirements.

Specifically, the TigerSHARC core includes 128 ea. 32-bit general purpose registers. This large number of registers allows C compilers sufficient flexibility to capitalize on the full potential performance of the architecture. In order to ensure data integrity, all registers are completely interlocked meaning that the programmer does not have to be cognizant of architecture delays. The hardware ensures valid data is used in computations. Additionally, all registers can be accessed via all addressing modes (orthogonal) and a deterministic delay (2 clock cycles) is achieved for all computational instructions. Lastly, the TigerSHARC architecture includes a Branch Target Buffer which holds the effective address of the last 128 branches or jumps. This feature alleviates the programming task of filling the instruction pipeline after branch instructions. If seen before, the architecture jumps to the next instruction in a single clock cycle.

ARCHITECTURAL FEATURES FOR HIGH-LEVEL LANGUAGE SUPPORT

- 128 General Purpose Registers
- All Registers Fully Interlocked
- General Purpose Integer ALUs for Addressing
- Branch Prediction
- No Hardware Modes
- Orthogonal Addressing Modes
- Assembly Language Support

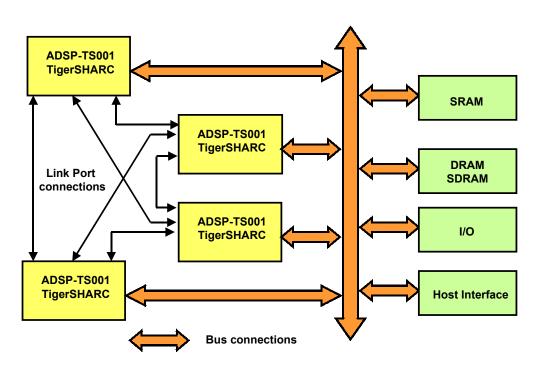
Figure 7.45

Figure 7.46 depicts one possible configuration of a TigerSHARC design in a multiprocessing implementation. Up to 8 ADSP-TS001 processors can communicate directly via the high speed 64-bit wide external bus interface. In this type of communication, a commonly used master-slave protocol is implemented which enables any two processors to communicate directly at any one time.

In addition to the primary external bus, a limitless number of processors can be connected via the ADSP-TS001 link ports. While offering more flexibility, link port connectivity provides lower per port bandwidth than the primary external bus interface. Again, all data transfers via the link ports are managed by a dedicated I/O processor and require no CPU intervention.

To summarize, the data I/O bandwidth of the link port (600MBytes/s) and external port (600MBytes/s) can be aggregated yielding an overall individual processor data bandwidth of 1200 Mbytes/s with 150MHz clock operation. Additionally, both the link port interface and the multiprocessor cluster interface are both completely glueless.

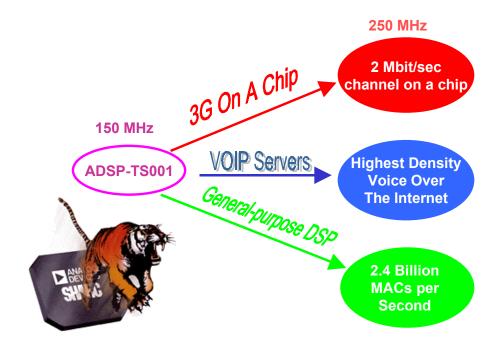
The ADSP-TS001 is the first member of a planned family of TigerSHARC-based products. Specifically, future members of the TigerSHARC family will contain optimized mixes of memory and peripherals to meet the requirements of specific target markets. These markets include third generation cellular base stations and VOIP (Voice Over the Internet Protocol) servers/concentrators. Additionally, process and design improvements will double the baseline performance of the general purpose TigerSHARC family members.



MULTIPROCESSING COMMUNICATION VIA LINK PORTS AND CLUSTER BUS

Figure 7.46

TigerSHARC ROADMAP





Comparing DSPs based solely on MIPS, MOPS, or MFLOPS does not tell the entire performance story. It is much more useful to compare DSPs based on their performance with respect to specific algorithms. The FFT and the FIR filter are popular benchmarks as well as the IIR biquad filter, matrix multiplications, division, and square root.

Figure 7.48 shows the benchmark performance of the ADSP-TS001 TigerSHARC operating on 16-bit fixed point data. Figure 7.49 shows its benchmark performance operating on 32-bit floating point data.

ADSP-TS001 TigerSHARC BENCHMARKS @ 150MHz 16-BIT PERFORMANCE

Algorithm	Execution Time	Cycles to Execute
256 Point Complex FFT (Radix 2)	7.3 µs	1100
50 Tap FIR on 1024 inputs	48 µs	7200
Single FIR MAC	0.93 ns	0.14
Single Complex FIR MAC	3.80 ns	0.57
Single FFT Butterfly	6.7 ns	1.0

■ 16-Bit performance -- 1200 MMACs/s peak performance

Figure 7.48

ADSP-TS001 TigerSHARC BENCHMARKS @ 150MHz 32-BIT PERFORMANCE

32-Bit performance -- 300 MMACs/s peak performance

Algorithm	Execution Time	Cycles to Execute
1024 Point Complex FFT (Radix 2)	69 µs	10300
50 Tap FIR on 1024 Input	184 µs	27500
Single FIR MAC	3.7 ns	0.55
Single FFT Butterfly	13.3 ns	2.0
Single Complex FIR MAC	13.3 ns	2.0
Divide	20 ns	3.0
Square Root	33.3 ns	5.0
Viterbi Decode(per Add/Compare/Select)	3.3 ns	0.5

DSP EVALUATION AND DEVELOPMENT TOOLS

The availability of a complete set of hardware and software development tools is essential to any DSP-based design. A typical DSP system design cycle is described below.

The first step in the process is to describe the system architecture. This would include such things as the type of processor, the peripherals (external memory, codecs, host processor, links), the configuration, etc. This information is placed in a file known as the *Link Descriptor File (or LDF)*.

The next step in the process is to generate the actual DSP code. This can be done using a higher level language (usually C or C++), the DSP assembly language, or a combination of both. DSP code developed in C must be *compiled* and *assembled* in order to generate the assembly language code. While programming in C is easier, the resulting assembly code which results after compiling is not as efficient as if the coding had been done in assembly language originally. For this reason, many DSP programmers do most of the programming in C, but use assembly language for the critical loops in the program. The Analog Devices' DSP assembly language is based on algebraic syntax and is relatively easy to use directly. The *linker* then generates an executable file.

The software must then be debugged using the *software simulator* in conjunction with an *evaluation board* such as the EZ-LAB evaluation board or perhaps a third-party card which plugs into a slot in the PC.

After the software is debugged using the evaluation board, it must be tested with the actual system target board (this is the board that you design with the DSP in your system). An in-circuit emulator, such as the EZ-ICE, interfaces with the target board, usually via a PCI or a JTAG interface port and connector.

The final step in the process is to generate the code required for booting the system using the *prom splitter*.

A summary of the tools available from Analog Devices is shown in Figure 7.50. Each one will be discussed in detail.

EZ-KIT Lites are basically DSP starter kit evaluation boards. In addition to the processor itself, these boards contain an ADC and a DAC (codec) which interfaces to the DSP over the DSP serial port. All necessary analog and digital support circuitry is contained on the boards. The options on the board are controlled over an RS-232 port connection to a PC as well as jumpers on the board. Windows 95/98/NT compatible software is supplied with the board. The software includes limited code generation tools including a limited feature set compiler, assembler, linker, prom splitter (loader), and Visual DSP debugger. Application examples such as DTMF generator, echo cancellation, FFT, simple digital filters, etc., are included as part of the software. The EZ-KIT Lite boards are primarily starter kit evaluation systems (and "lite" on the wallet!).

ADI DSP DEVELOPMENT TOOLS

- **■** EZ-KIT Lite[™] Evaluation Boards
- Emulators
- Integrated Development Environment (IDE) Software, VisualDSP[®] and VisualDSP++™
 - Assembler, Linker, PROM Splitter, HIP Splitter, Simulator, Compiler, Debugger
- Extensive Algorithm Libraries
- Factory, Field, and WWW Support
- Seminars
- ADI and Third Party DSP Workshops
- ADI DSP Collaborative[™] Third Party Support

Figure 7.50

EZ-KIT LITE'S[™] FOR ANALOG DEVICES' DSPS

- The EZ-KIT Lite[™] is a stand-alone (desktop) system that connects to a PC running on Windows
- The EZ-KIT Lites provide:
 - A cost effective method for initial evaluation of the capabilities of ADSP-series DSPs.
 - A powerful development environment for a variety of general purpose applications.
- Target market:
 - First time DSP users
 - First time ADI DSP users
 - Existing ADI DSP users implementing new designs
 - Existing ADI DSP users upgrading to faster devices for current designs

ADSP-2189M EZ-KIT LITE[™]

- Hardware Features
 - ADSP-2189M 75 Mips processor
 - AD73322L Stereo codec
 - DSP-programmable CODEC gain
 - 2 Mbit or greater boot protected Flash EPROM.
 - RS-232 PC to EZ-Kit Lite interface
 - Selectable Host vs. Full Memory mode implemented via dip switch
 - ADSP-218x EZ-ICE emulator port connector
 - Expansion connector includes all signal I/O plus 5V, 3.3V, 2.5V, and GND connections
 - LED indicators for master power, RS-232 interface, and one PF I/O
- Software Features
 - Windows 95/98/NT-4.0 PC host support
 - VisualDSP[®]: Limited feature set compiler, assembler, linker, prom splitter (loader), VisualDSP debugger interface
 - Application Examples: DTMF Generator, Echo Cancellation, FFT, etc.(similar to 2181 EZ-KIT Lite)
 - Email Support

Figure 7.52

ADSP-21160M EZ-KIT LITE[™]

Hardware Features

- ADSP-21160M SHARC processor
- AD1881 16-bit Stereo AC'97 SoundMAX Codec
- EPROM flash memory (2 Mbit)
- JTAG header
- Support for ADSP-2116x family of processors
- 64K x 64 bit SBSRAM
- Enhanced parallel port
- CE compliant
- Software Features
 - Support for Win95, Win98 and WinNT
 - Evaluation suite of VisualDSP++[™] : compiler, assembler, linker, prom splitter (loader), VisualDSP debugger interface. VisualDSP limited to use with EZ-KIT Lite hardware
 - DEMONSTRATIONS: DFT.dxe, BP.dxe, Pluck.dxe, Primes.dxe, Tt.dxe



ADSP-21065L EZ-KIT LITE™

- Hardware Features
 - ADSP-21065L SHARC DSP running at 60MHz
 - Full Duplex, 16-Bit Audio Codec
 - RS-232 Interface with UART
 - JTAG Emulation Connector
 - Expansion via MAFE+ Connector
- Software Features
 - Support for Win95, Win98 and WinNT
 - Evaluation suite of VisualDSP++™: compiler, assembler, linker, prom splitter (loader), VisualDSP debugger interface.
 VisualDSP limited to use with EZ-KIT Lite hardware
 - Demonstrations: Fast Fourier Transform (FFT), Discrete Fourier Transform (DFT), Band Pass Filter, Pluck String Themes, Talk Through

Figure 7.54

The final step in the DSP system development is the debugging of the actual system, or "target" board. The Analog Devices' in-circuit emulator, EZ-ICE, interfaces with a connector on the target board for use in final system hardware and software debugging. Examples are shown in Figures 7.55 through 7.58. Figure 7.56 shows the Apex-ICE which interfaces to the target board via a JTAG connector which in turn interfaces to the SHARC DSP. A USB port connector is used to interface the emulator to a PC. Other in-circuit emulators are available which interface to ISA, PCI, RS232, and Ethernet ports.

EZ-ICE® FOR THE ADSP-218x DSP FAMILY

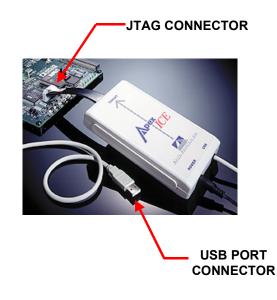
- Serial port interface, printed circuit board and 14-pin header
- Controls equipment for testing, observing, and debugging a target system
- 6 foot cable
- Hardware switch to accommodate of 2.5V, 3.3V, and 5V
- Shielded enclosure to cover bare circuit board
- Performance increase via faster data transfer



Figure 7.55

APEX-ICE[™] USB EMULATOR

- Universal Serial Bus (USB)based emulator for Analog Devices JTAG DSPs
- First portable solution for Analog Devices JTAG DSPs
- Small hand-held unit
- Small diameter cable, 5 meters in length, for hard to reach targets
- Power provided externally



TREK-ICE[™] ETHERNET EMULATOR

- Network hosted mini-tower emulator with 10-Base-T port
- Installs to a LAN as easy as a laser printer
- Remote debugging between either PC or SUN workstation debug host (client) and the target DSP system
- Rugged high-speed 3V/5V pod
- Flexible emulator cable (1.5m)



Figure 7.57

SUMMIT-ICE[™] PCI EMULATOR

- **32-bit PCI interface add-in card**
- Four inch, flexible shielded target board cable for easy access to a 14-pin JTAG header
- Embedded ICEPAC technology provides a rugged and reliable solution
- Remote 3V/5V JTAG Pod with extended, shielded cable (1.5 m)
- Windows 95 & NT PNP



VISUALDSP[®] AND VISUALDSP⁺⁺

New development software for Analog Devices' DSPs is written in easy to use $VisualDSP^{\textcircled{R}}$ and VisualDSP ++ which is Windows 95/98/NT compatible. VisualDSP is a completely integrated development environment which uses an algebraic syntax assembler and an optimized C compiler. Multiprocessor environments can be simulated and debugged as well. VisualDSP++ provides C++ language support.

VisualDSP versions currently exist for the ADSP-218x and ADSP-219x families as well as the SHARC family of DSPs.

A "test drive" CDROM is available with a limited license for evaluation purposes.

In addition to the tools and support functions described thus far, Analog Devices' DSP Collaborative consists of over 80 companies who provide a range of products and services to make the DSP design task easier. Over 30 companies support the 16-bit ADSP-21xx family, and over 50 companies support the SHARC DSP family. A directory of the collaborative can be found at:

http://www.analog.com/industry/dsp/3rdparty/index.html

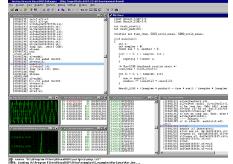
Further information about Analog Devices' DSP tools can be found at:

http://www.analog.com/dsp/tools

SOFTWARE DEVELOPMENT ENVIRONMENT

- VisualDSP® and VisualDSP++
 - Debugger front-end
 - Integrated development environment (IDE)
 - Algebraic syntax assembler
 - Cycle-accurate instruction level simulator
 - Optimizing ANSI C compiler with inline assembly
 - Math, DSP, and C runtime libraries
 - Sophisticated multiprocessor linker
 - Intelligent loader
 - ADSP-218x, 219x: Windows 95, 98, NT, 2000 Compatible
 - SHARC: Windows 95, 98, NT, 2000 Compatible





VisualDSP® 7.0 FOR ADSP-218x AND ADSP-219x

- Hosted on Windows 95, Windows 98, Windows NT 4.0 with SP3 or later
- ADSP-219x simulator target supports the 219x core
- ELF/DWARF toolset, including compiler with classical and processor-specific optimizations
- Tcl command line scripting language
 - Support automated test of DSP system
- 21xx Object Translator
- Peripheral Code Wizard supporting 218x and 219x peripherals

Figure 7.60

VisualDSP++[™] FOR SHARC® DSPs

- Supports Windows 95, 98, NT, 2000
- ELF/Dwarf-2 file format enables effective debug operation
- Pre-processor for linker/assembler separate from compiler
- Fast ICE stepping 0.9 sec. per step
- MultiProcessor (MP) support
 - Synchronous run, step, and halt
- Tcl command line scripting language
 - Support automated test of DSP system
- Statistical Profiling
- C++ capabilities
- Graphical Plotting

TigerSHARC® DEVELOPMENT TOOLS

- The TigerSHARC architecture is supported by a robust set of simulation, code generation, and debug tools that includes:
 - VisualDSP Integrated Development Environment
 - Simulator, Assembler, Loader, Debugger, and Compiler
 - DSP & Math Libraries
 - Emulators All present ADI JTAG emulators support the TigerSHARC DSP family
 - EZ-KIT Lite
- Successive VisualDSP revisions will continuously increase functionality.

Figure 7.62

VisualDSP TEST DRIVE

- The test drive is a 30-day evaluation of VisualDSP's full package. It does not include a tutorial. The new test drive is a full version of VisualDSP and contains pdf.'s of the VisualDSP manuals.
- The customer is presented with the test drive CD. The customer than proceeds to the Analog Devices DSP Tools website, clicks on Test Drive Registration

www.analog.com/industry/dsp/tools/test_drive.html and registers online. After they complete the registration process they will receive a serial number immediately that will allow them to use the test drive. The test drive will expire 30-days from the install and they will not be able to register the test drive again.

- The SHARC VisualDSP test drive is now available, SAP part # VDSP-SHARC-PC-TEST.
- The VisualDSP TigerSHARC test drive will be available Summer 2000
- The VisualDSP ADSP-218x/219x test drive will be available in September 2000

ADI DSP COLLABORATIVE - WHAT IS IT?

- Over 80 Companies that provide a wide range of products and services to make your design challenge easier
- Architecture Coverage
 - Over 30 Companies support 16-bit, ADSP-21xx Family
 - Over 50 Companies support SHARC[®] DSP Family
- Over 400 Products from the Following Categories:
 - Algorithms

- Emulators
- Real-Time Operating Systems
- Debuggers
- MATLAB[®] DSP Support
- Focused Applications :
 - Audio
 - Digital Radio

 - Medical Instrumentation/Imaging
 - Military/Aerospace

- Hardware Development Boards
- Graphical S/W Programs
- Consulting Services
- Motor/Motion Control
- Radar/Sonar

- http://www.analog.com/industry/dsp/3rdparty/index.html

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- 2. C. Britton Rorabaugh, **DSP Primer**, McGraw-Hill, 1999.
- 3. Richard J. Higgins, **Digital Signal Processing in VLSI**, Prentice-Hall, 1990.
- 4. Ethan Bordeaux, Advanced DSP Performance Complicates Memory Architecures in Wireless Designs, Wireless Systems Design, April 2000.
- 5. **DSP Designer's Reference (DSP Solutions)** CDROM, Analog Devices, 1999.
- 6. **DSP Navigators: Interactive Tutorials about Analog Devices' DSP Architectures** (Available for ADSP-218x family and SHARC family): http://www.analog.com/industry/dsp/training/index.html#Navigator
- 7. **General DSP Training and Workshops**: http://www.analog.com/industry/dsp/training

The following DSP Reference Manuals and documentation are available for free download from: http://www.analog.com/industry/dsp/tech_docs.html

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- 9. ADSP-2100 Family EZ Tools Manual.
- 10. ADSP-2100 EZ-KIT Lite Reference Manual.
- 11. Using the ADSP-2100 Family, Vol. 1, Vol. 2.
- 12. ADSP-2106x SHARC User's Manual, 2nd Edition, July, 1996.
- 13. ADSP-2106x SHARC EZ-KIT Lite Manual.
- 14. ADSP-21065L SHARC User's Manual, Sept. 1, 1998.
- 15. ADSP-21065L SHARC EZ-LAB User's Manual.
- 16. ADSP-21160 SHARC DSP Hardware Reference.

DSP HARDWARE