

A functional block diagram of how the SR-830 DSP LIA works is shown in the figure below:

The dual-channel SR-830 DSP LIA is a *digital* lock-in amplifier – this means that the voltage waveform of the analog input signal is sampled at a 256 *KHz* digitization rate using a 16-bit analog-to-digital converter. A Phase-Locked Loop (*PLL*) circuit is used to lock the phase of the reference sine-wave (or in the form of a *TTL* digital signal) to an internal 20-bit *digital* oscillator. {Or, the internal 20-bit digital oscillator of the SR-830 DSP LIA can be used as the reference sine-wave, programming its amplitude and frequency settings via the GPIB (General Purpose Instrumentation Bus) interface to a PC. The LIA's Digital Signal Processor (*DSP*) then carries out *all* of the necessary mathematical operations (multiplication, 90° phase shift, low-pass digital filtering) discussed above, in the *digital* domain. This approach is *vastly* superior in terms of improved stability, reduced noise, overall performance & flexibility/versatility in comparison to the performance of the *analog* lock-in amplifiers of yesteryear... it is truly a powerful device, one which has been used in countless laboratory settings, and in countless physics, engineering, biology, ... applications!

Note that the SR-830 *DSP LIA* can also be used to analyze the <u>higher</u> harmonics associated with <u>*polyphonic*</u> complex sound fields, consisting of a <u>hierarchy</u> of {<u>*precisely*</u>} integer-related overtones (up to n = 99!).