



Spectral Processing for GPS Time Stamped Signals

Application Note 085



by James Zhuge, Ph.D. - Chief Executive Officer
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Abstract

The Ground Recording System (GRS) is a new product developed by CI that can accurately time stamp acquired digital data using Global Position System (GPS) technology. The time marks can be as accurate as 100 nanoseconds. Using this time stamping technology, we developed the theory and algorithms to calculate the auto and cross spectrum between measurement channels on separate **data acquisition units that can be distanced thousands of miles apart.**

Other traditional time sources, such as IRIG-B, can be used in the same manner as a GPS time source. In this paper, we mainly address the technical issues raised in using a GPS time source since it is a new technology in our industry.

Traditionally, measuring the transfer function between any two sensors requires that all ADCs (Analog to Digital Converters) are accurately synchronized through a hardware connection. Synchronizing the sampling clocks of all ADCs is accomplished when all the data acquisition units are located nearby. Crystal Instruments has developed Ethernet based time synchronization technology that can extend the cable length up to 300 feet. The ADC clock synchronization will become difficult and unreliable beyond 300 feet. The technology presented in this paper provides an alternative approach. Instead of synchronizing the ADC sampling clocks, this new technique only requires accurate time stamping to the sampled data points.

This technique allows measurement channels that are separated by a large distance to synchronously acquire data without any connected wires.

Most typical FFT analyzers always compute the following spectra: auto power spectra of all measurement

channels, cross-spectra of any pair of channels, transfer function (or FRF, the Frequency Response Function) of any pair of channels, coherence functions of any pair of channels, and phase spectra of any pair of channels. Once the auto power spectra and cross-spectrum of a pair of channels are computed, the other signals can be derived. Since the auto-power spectra computation is very simple and requires no cross-term information, the only question remaining is how to compute the cross-spectrum of any pair of channels that are from different GRS units and are accurately time stamped.

Signals are received from separate GRS units, so a new algorithm was developed to correct the phase mismatch between any pair of FFT spectra based on certain calculations of time stamps. Typical tests indicate that the GRS can achieve the following specification for the phase estimation of the cross-spectral calculation:

Phase match between two GRS units:	$\pm 2^\circ$ (degree) at 40 kHz
	$\pm 1^\circ$ (degree) at 20 kHz
	$\pm 0.5^\circ$ (degree) at 2 kHz

This is an extraordinary result. While many companies claim to use a GPS time base for their data acquisition, we have not seen any other companies claiming to achieve such an outstanding cross-term spectral analysis performance.

Potential Applications

This technology has many potential applications. Essentially, any structure or acoustic measurement that requires multiple sensors distributed across a large distance can now use the GRS to perform the measurement. After obtaining the measurement, the cross-channel terms can be calculated with greater accuracy.

1. Sound boom measurement for aircraft

The contract awarded to CI by NASA intends to use hundreds of GRS units spread across 10,000 square miles to simultaneously take measurements. While the contract did not specify the details of the time stamped signal processing method described in this document, the technology can certainly be used to analyze the sound boom effect with much higher time accuracy.

2. Measure the vibration mode of large bridges

Using only two GRS units and by

roving the measurement points, it is possible to measure the operating deflection shape of a very large bridge. All we need is a sequence of FRF measurements in pairs.

3. Vibration mode of large vehicles

The GRS is ideal for measuring the vibration mode of ships, aircraft, large trucks, bullet trains, railways, large cranes, and other large structures.

In theory, two GRS units can acquire data across the earth's diameter, and we can calculate the transfer function between the measurements of these two systems. But we have not seen a need for it.

The only limitation of this technology is that the data acquisition systems need to receive GPS signals. Without GPS signals, the time stamps are not available.

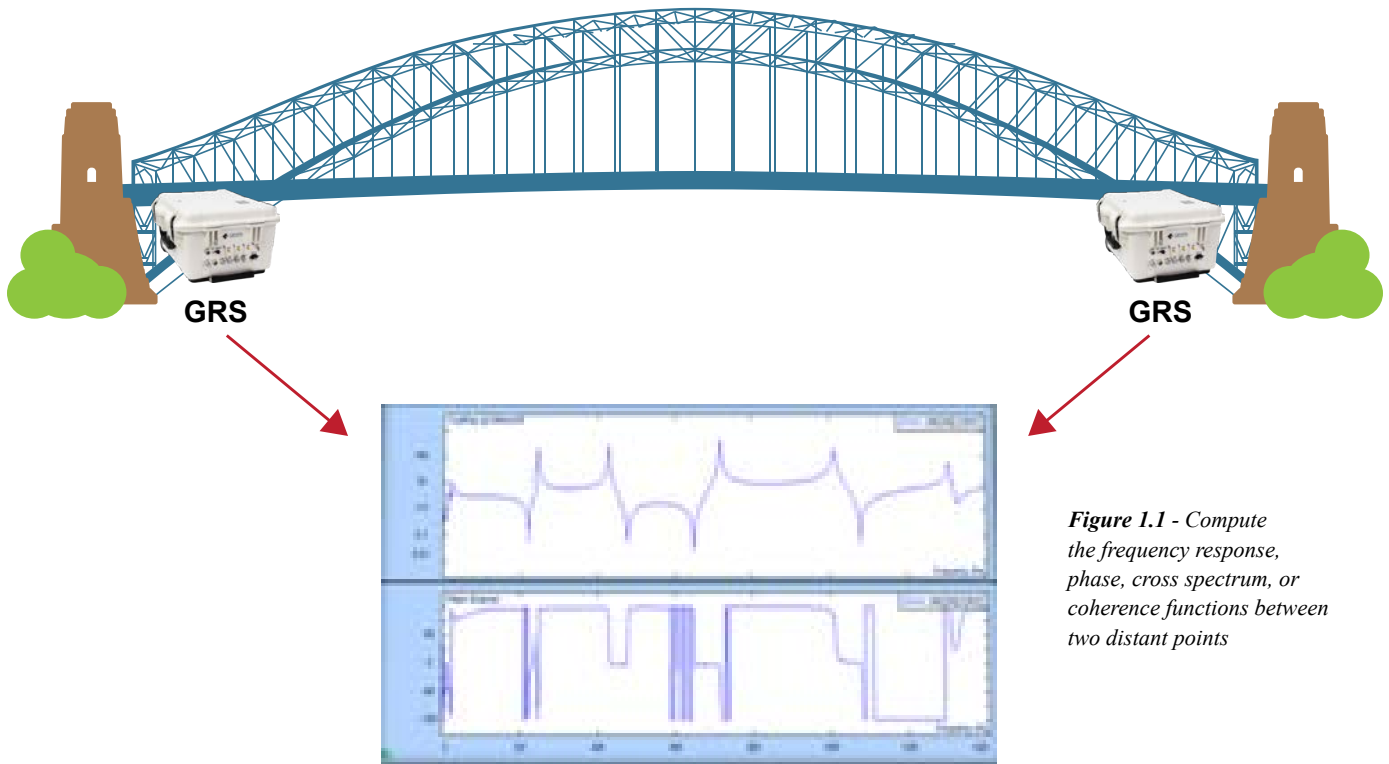


Figure 1.1 - Compute the frequency response, phase, cross spectrum, or coherence functions between two distant points

Analysis of GPS Time Stamps

Technology background of time stamping

The Global Navigation Satellite System (GNSS) installed in the GRS delivers high integrity, precision timing for demanding applications across the world. Support for BeiDou, GLONASS and Galileo constellations - in addition to GPS (Global Position System) - enables compliance with national requirements. Enhanced sensitivity and concurrent constellation reception extends coverage and integrity to challenging signal environments. Survey-in and fixed-position navigation reduce timing jitter, even at low signal levels, and enable synchronization to be maintained with as few as one single satellite in view.

Due to the popular use of the term GPS (Global Position System) in history, this document simply adopts the usage of GPS instead of GNSS.

In a data acquisition system where the time accuracy of the sampling clock is not very demanding, say no better than seconds, the time base can be derived from the “computer time”. This can be set manually, through a time server on a network, or through the internet. When time accuracy of the sampling clock demands millisecond resolution, the digital input paths of the data acquisition system - especially its ADC - has to be designed with control from a more accurate time base, such as GPS or IEEE 1588 PTP (precision time protocol). This is necessary to calculate the cross-channel spectrum, or any signal property related to the time delays between all measurement signals.

If the ADC sampling clocks cannot be accurately synchronized between different hardware units, it is possible to achieve similar

signal processing results if all recorded data is accurately time stamped. Applying time stamps and synchronizing the ADC clocks are two different strategies. CI believes that with accurate time stamping, the users should be able to achieve the same measurement goals, including calculating the cross-spectra between all measurement channels.

Multiple GRS data acquisition systems can acquire data at the same time while they are physically spread out over hundreds of miles. These units do not share any direct hardware connections except that each will receive GPS signals. Accurate time

stamping technology is implemented with the GPS time base to allow the acquired signals to line up in Crystal Instruments’ post processing software, Post Analyzer (PA).

Global Positioning System, GPS, consists of 24 satellites revolving around the earth at every 12 hours. Each of these satellites has an atomic clock onboard with extremely good accuracy. The GPS satellites continuously transmit their coordinates in space along with a time message on a 1.575 GHz carrier frequency. The GPS receiver used in the GRS claims to achieve ± 60 ns time accuracy at 99% of the time.

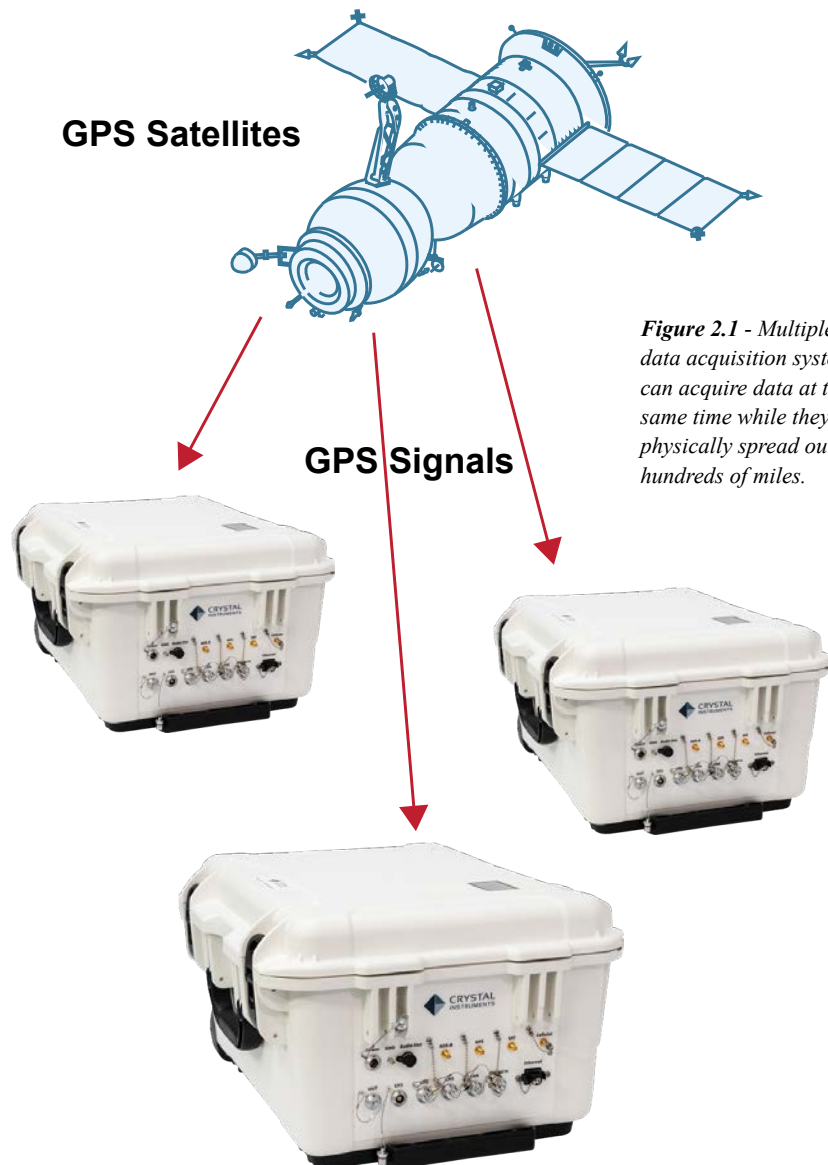


Figure 2.1 - Multiple GRS data acquisition systems can acquire data at the same time while they are physically spread out over hundreds of miles.

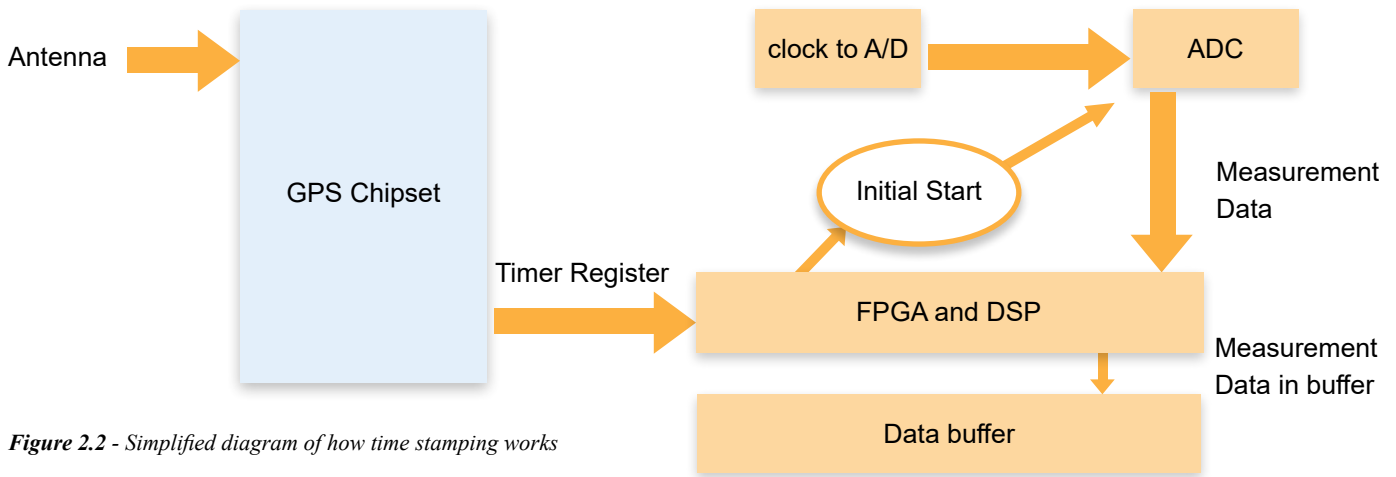


Figure 2.2 - Simplified diagram of how time stamping works

Clock Drift Analysis

Clock signals, timing, and location signals can be derived from the GPS chip signals inside each GRS unit. A real-time, zero latency hardware logic is implemented to time stamp the A/D sampling clock with the measured GPS time base.

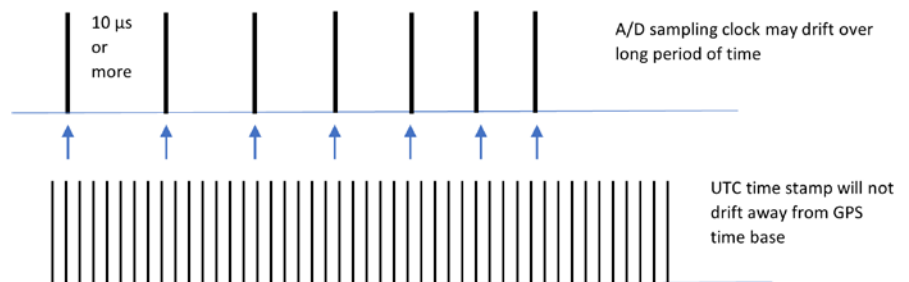


Figure 2.3 - Diagram shows the ADC clock drift in concept

Figure 2.2 provides a simplified diagram of how time stamping works in the GRS.

In the GRS, a time register is derived from the GPS receiver chip that is directly accessible by the FPGA and the processor. The GPS timing accuracy is better than ± 60 ns with 99% probability according to the specifications of the GPS chipset. The time stamping method described in the preceding section results in time stamping accuracy within the sub-microsecond range, typically less than 100 ns.

Given such an accurate technique of applying time stamps, we are able to analyze the error of actual sampling clocks vs. the GPS time base, which is described as follows.

An ADC requires a clock with a specific oscillator frequency to drive its sampling. The diagram in Figure 2.3 shows the ADC clock drift in concept. Here the ADC clock has a period of more than $10 \mu\text{s}$ (102.4

$\text{kHz} \leftrightarrow 10 \mu\text{s}$ sampling). The drift might be caused by the environment, especially changes in temperature.

Besides the short term *drift*, the ADC sampling clock - which is driven by an internal oscillator in the GRS - may have an inherent *offset* compared to oscillators on other GRS units, or compared with the atomic clocks used by GPS time. The combination of offset and drift produces a *bias* of the clock. For a recording time of a few minutes this bias may not be a major problem. However, if the recording time is as long as hours or days, this sampling clock bias may cause sampled data from different units to have mismatches when comparing time stamps.

$$\text{Bias error} = \text{short term drift} + \text{offset}$$

Suppose the oscillator that drives the A/D sampling has a stability of 50 ppm (part per million). After one hour the clock bias will be the following:

$$(0.000050 \text{ bias}) * (3600 \text{ seconds per hour}) = 180 \text{ ms bias error per hour}$$

In the worst case scenario, the time clock of the sampled data will vary by a factor of 180 ms per hour. This situation occurs when all oscillators on different GRS units are running without clock constraints, such as periodic synchronization through a GPS signal.

In order to discuss the timing issues with clarity, a few terms used in this document are introduced:

Nominal Sampling Rate is the sampling rate labeled by the product specifications, for example, 64 kHz, 102.4 kHz etc. They are not accurate because of the bias and drift of the ADC clocks.

Nominal Time is the time of acquired signal samples calculated based on a starting time, number of samples, and nominal sampling rate.

Measured Time is a series of time stamps taken from the GPS clocks corresponding to the samples of acquired signals. If the GPS works properly, the Measured Time is a much better time base than Nominal Time.

Corrected Sampling Rate is a single value for the sampling rate used in the entire duration of a recording. It is calculated by removing the offset between local clocks against that of GPS based on the Nominal Sampling Rate, duration, and time offset.

In other words, the **Corrected Sampling Rate** is the calculated sampling rate by removing the first order error between the local clocks and GPS time base.

Figure 2.4 provides an example of the format used by the GRS to display recorded time stamps .

When drawn in a plot using *(Measured-Nominal) vs. Nominal*, it will look similar to the example provided in Figure 2.5.

In this example of roughly 500 seconds of measurement, the nominal time is off by 5 milliseconds against the GPS time clock. In this particular case, the local clock is slightly slower.

The 5 milliseconds error from 500 seconds of measurement is within the range of expected error of the oscillator. But it is too large to conduct signal processing when cross-channel computation is needed. For vibration and acoustic applications, the time base difference between the data samples of any two channels must be within a few microseconds of each other.

Since we know the error between nominal time and actual measured is mainly an offset instead of random drift, this offset error can be corrected

	Nominal Time(REC0078.ts)	Actual Time(REC0078.ts)
▶ 0	00:00:00.000.000.000	00:00:0
1	00:00:03.000.000.000	00:00:02.999.606.7...
2	00:00:06.000.000.000	00:00:05.999.213.5...
3	00:00:09.000.000.000	00:00:08.998.885.8...
4	00:00:12.000.000.000	00:00:11.998.492.6...
5	00:00:18.000.000.000	00:00:17.997.706.2...
6	00:00:24.000.000.000	00:00:23.996.985.3...
7	00:00:30.000.000.000	00:00:29.996.198.9...
8	00:00:48.000.000.000	00:00:47.993.905.1...
9	00:00:57.000.000.000	00:00:56.992.791.0...
10	00:01:03.000.000.000	00:01:02.992.004.6

Figure 2.4 - GRS records time stamps and displays them in the format shown here.

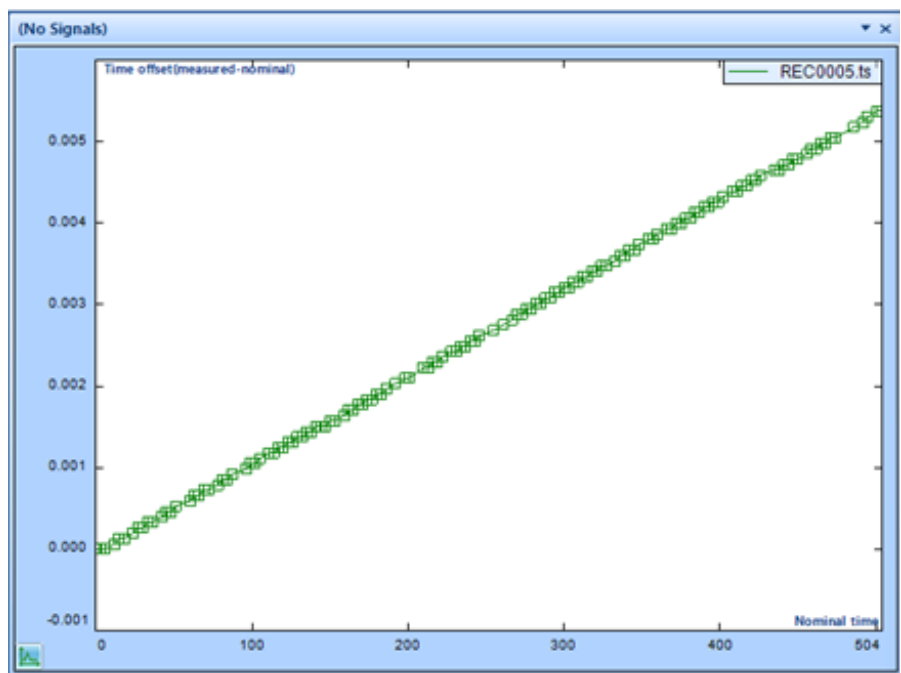


Figure 2.5 - drawn in a plot using *(Measured-Nominal) vs. Nominal*

after taking the measurement. The technology developed to create the GRS first applies a correction to the sampling rate, then uses the signals with the corrected sampling rate in post processing. The following describes the correction process with more details:

1) Calculate the Time Offset as shown

in the previous plot for the entire duration of the measured signals, 504 seconds in this example.

2) Based on the duration of measurement and the time difference between that of nominal and GPS time stamps, the actual measured sampling rate is calculated as:

Corrected Sampling Rate = Nominal sampling rate / (1.0 + Time Offset/ Duration)

For example, if the Nominal Sampling Rate is 64 kHz, the Time Offset is 0.005 second and Duration is 504 seconds,

Corrected Sampling Rate = 64 kHz / (1.0 + 0.005/504.0) = 63.999365 kHz

3) This Corrected Sampling Rate is computed by removing the first order error from the nominal sampling rate. It did not take the sampling rate fluctuation into consideration during this time period.

4) Since we know the signals acquired on this unit come from a slower clock of 63.999365 kHz, we are able to adjust it in the display and post processing algorithm.

The following is an example of how the corrected sampling rate can help in data processing.

The plot in Figure 2.6 shows that when an impulse event is recorded by two data acquisition systems with slightly deviated sampling clocks, the signals will be off by a certain duration even with the same starting time-base.

But after a correction against the GPS time base, the sampled points can be lined up on the plot, as demonstrated in Figure 2.7.

After the adjustment, all classical signal processing methods such as FFT, FRF (Frequency Response Function), Coherence Function etc., can be applied using the post processing software.

The next question asks how accurate the Corrected Sampling Rate is. The post processing software is able to display the clock fluctuation after the first order correction. The two graphs

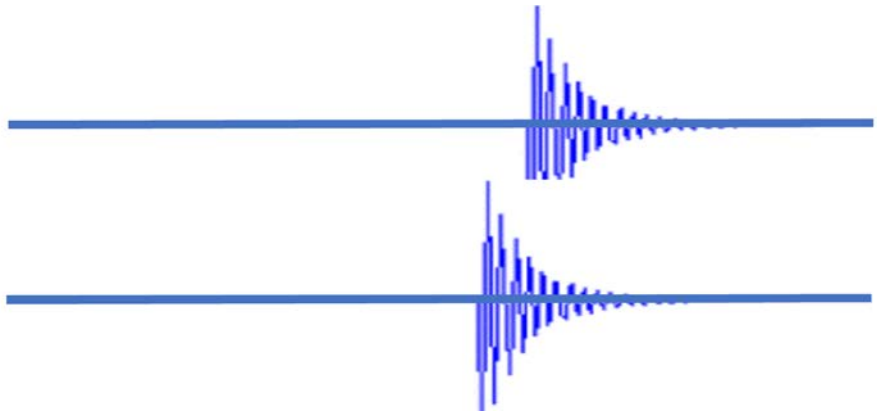


Figure 2.6 - an impulse event is recorded by two data acquisition systems with slightly deviated sampling clocks will have signals off by a certain duration even with the same starting time-base

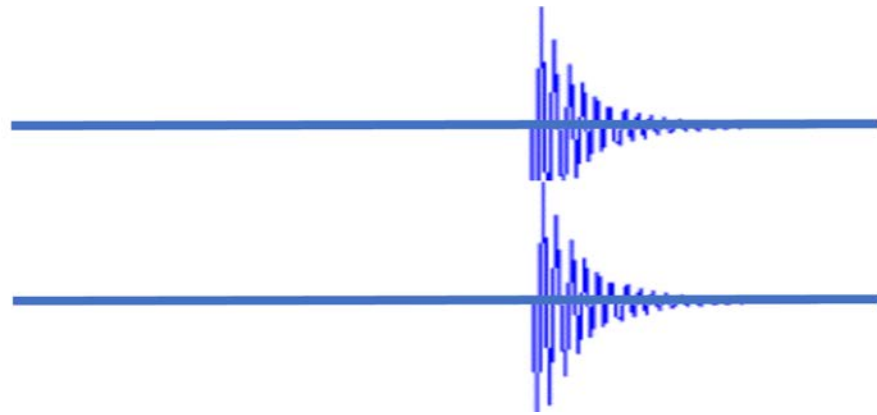


Figure 2.7 - after a correction against the GPS time base, the sampled points can be lined up on the plot

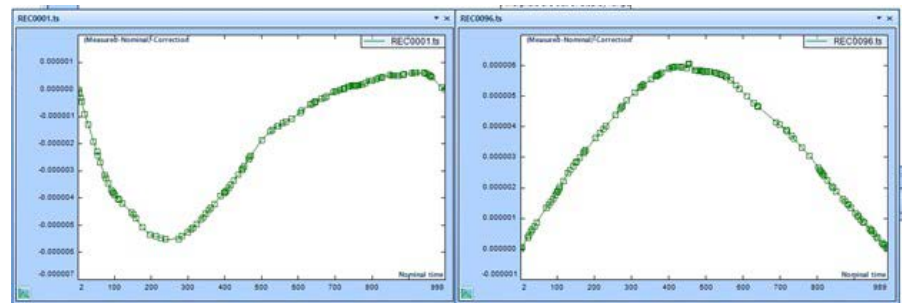


Figure 2.8 - displays the “drift” of the time stamps after the correction is applied

provided in Figure 2.8 display the “drift” of the time stamps after the correction is applied.

Both vertical and horizontal are displayed in units of seconds. In these graphs, the quantity displayed vertically is:

(Measured – Nominal) – Correction

The **Correction** is a straight line

drawn between the two time stamps, the very first and last time stamps of the entire recording. Both signals show that the fluctuation range is roughly 6 μs while the recording time is roughly 1000 seconds.

This fluctuation will be further corrected when a more sophisticated clock correction algorithm is used, which is already implemented in the Crystal Instruments PA software.

Calibrate the fixed offset of time stamps

The ADC and filtering processing chain in the GRS data acquisition system will introduce a fixed time bias to the time stamps. This time bias is governed by the sampling rate.

The diagram provided in Figure 2.9 shows the three areas that will contribute to this time bias error.

ADC filter delay: the built-in FIR filter in the sigma-delta ADC has a certain delay effect. The time delay depends on the filter type that the hardware is initialized to, and the sampling rate. Since the filter type will never change on a GRS unit, the only accessible parameter is the sampling rate. The ADC sampling rate is selectable from 8 kHz to 102.4 kHz in multiple stages. Each sampling rate setting will have a fixed amount of time delay for ADC.

FPGA temporary buffer (part of propagation delay): The time stamps are attached to the input sampled data in the FPGA. The FPGA may have certain temporary buffers that may cause a certain amount of time delay. The length of these temporary buffers is fixed. Therefore, the time delay caused by FPGA is also fixed.

Decimation filter on DSP (filter delay): at a sampling rate lower than 8 kHz, one or multiple FIR type of decimation filters will be inserted at the DSP level. The decimating filters will cause a fixed time offset bias. The only determining factor is the user selection of a sampling rate.

In summary, given a sampling rate setting from the user interface, the time bias error is a fixed constant number across all GRS input channels. To correct this bias error, it is simply a matter of adding a correction table to all time stamps with the input parameter of a

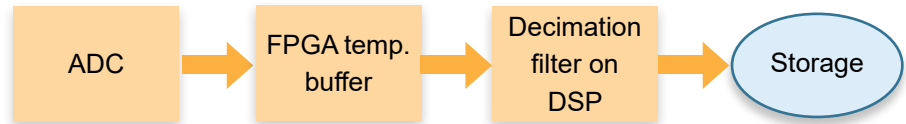


Figure 2.9

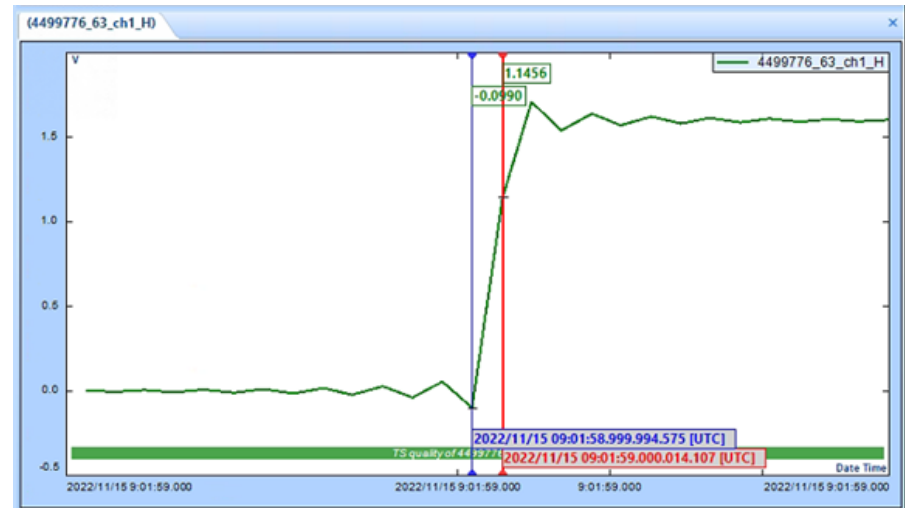


Figure 2.10

sampling rate. The recorded data will have time stamps with the bias removed.

The signal plot example provided in Figure 2.10 shows how after the time offset bias is completely removed, the 1 pps (pulse per second) GPS signal is fed into one of the input channels.

The data is sampled at 51.2 kHz. The time interval between any two discrete points is about 20 μ s. The two cursor values show the UTC time stamps at two discrete points. The timing of rising edge of the 1 pps signal is from the GPS chip mounted on the GRS. It can be traced back to the National Institute of Standards and Technology (NIST) time source. The rising edge of the pulse shown in this example is at the following time:

2022/11/15 09:01:59 [UTC]

The time at the blue cursor is about 6 μ s less than this “integer” seconds while the time at the red cursor is roughly 14 μ s more than this “integer” seconds.

After the bias removal, the absolute value of time stamps will have no offset error.

Determining the interval of time stamps

In the GRS system, the interval of time stamps is chosen as 5 seconds. In roughly every 5 seconds a time stamp is taken. The time stamps and geographic location data values are saved into a separate file (.TS file). The file will be encrypted and saved into the SD cards.

The choice of 5 seconds as a time stamp interval is not arbitrary. We chose this length based on the following factors:

1. Considering the drift error of the local ADC clock is 10 ppm, we calculate the biggest possible time error of an ADC clock for a time stamp interval as larger than 5 seconds.
2. Using the cross-spectrum analysis method described below, we analyze the largest possible phase shift when increasing the interval of time stamps.

3. The interval should not be unnecessarily too small in order to save memory space.

It is found that if we increase the time interval of time stamps to beyond 100 seconds, the phase measurement using the cross spectral method will show an unacceptable deviation. But as long as the time interval of time stamps is less than 60 seconds, the phase measurement of cross spectrum is always good. Therefore, a 5 second interval is a very conservative choice.

When time stamps are lost during data acquisition

The GRS has independent ADC clocks that drives the ADC sampling and a system clock that keeps the local time and calendar. When GPS signals are not available, the time stamps and geographic location information will not be available. But the GRS can still conduct normal data acquisition operations.

The GPS timing base, if available, will be assigned to the system clock each time the GRS is booted up. This means that the system clock will be frequently adjusted based on the GPS time base.

If the GPS signals are not available when the system is booted up and the data acquisition starts:

1. The GRS system will assign its previously saved location coordinates to the ATRX file header.
2. The GRS system will assign its system clock value, which is less accurate than that of GPS, to the ATRX file header as the starting time.
3. The .TS file will be filled with a mark of “****TS lost****” since time stamps are not available for use. Figure 2.11 provides an example of the time stamp display with TS lost.

If during the initial stage of data

	X Nominal time (Seconds)	MergedSig_ch1 (Y)	MergedSig_ch2 (X)	MergedSig_ch2 (Y)
172	870.0000000000	0.0000410688	865.0000000000	-0.0000220741
173	875.0000000000	0.0000415713	870.0000000000	-0.0000221986
174	880.0000000000	0.0000420768	875.0000000000	-0.0000223310
175	885.0000000000	0.0000425673	880.0000000000	-0.0000224435
176	890.0000000000	0.0000430738	885.0000000000	-0.0000225849
177	895.0000000000	0.0000435603	890.0000000000	-0.0000227074
178	900.0000000000	0.0000440478	895.0000000000	-0.0000228298
179	905.0000000000	0.0000444653	900.0000000000	-0.0000229152
180	**** TS lost ****		905.0000000000	-0.0000229727
181	920.0000000000	0.0000460418	910.0000000000	-0.0000230211
182	**** TS lost ****		**** TS lost ****	
183	3320.0000000000	0.0000826384	925.0000000000	-0.0000232275
184	3325.0000000000	0.0000826979	930.0000000000	-0.0000232179
185	3330.0000000000	0.0000825694	935.0000000000	-0.0000232593

Figure 2.11

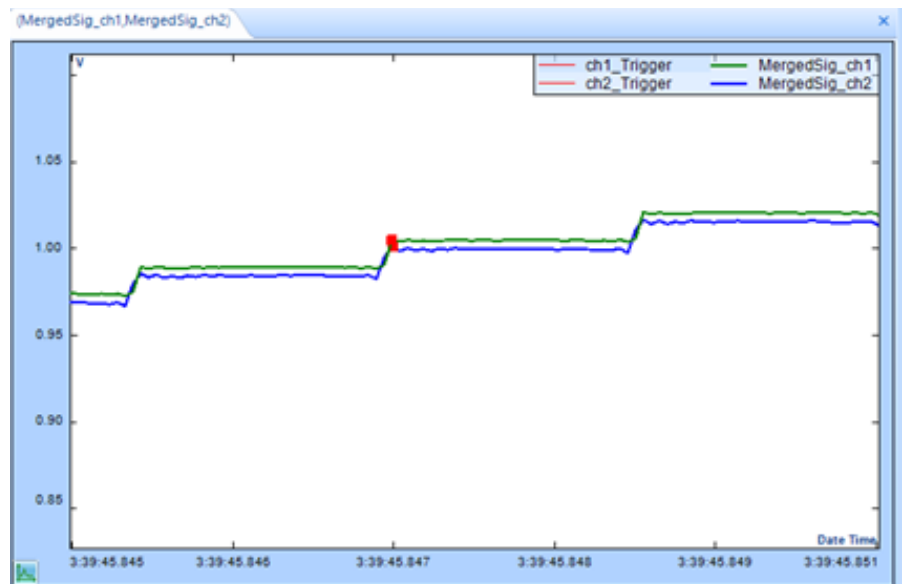


Figure 2.12

acquisition, the GPS signals are not available but the GPS signal returns during the operation:

1. The .TS file will be filled with a mark of “****TS lost****” in the duration that time stamps are not available to use.
2. When GPS signals are good, the GRS system will start to record the location and time stamps into the .TS file
3. The GRS system will recalculate

the acquisition starting time based on GPS information and save it to the ATRX file header.

Our experience indicates that if the GPS signals are lost for about 60 seconds or less, the quality of time stamps will not affect the measurement. But if the GPS signals are lost for longer than 100 seconds, the cross spectral analysis will not achieve good phase quality.

An example of time alignment

The following is an example

demonstrating how the signals are plotted after a sampling rate correction is applied using attached time stamp signals.

Original file name: REC0058, REC0131. These are recorded on two GRS units. When they are recorded, a source signal is fed into the input ends of both units using a T-splitter port. Both GRS units have installed GPS receivers to communicate with GPS satellites. The recording duration is about 40 minutes.

Nominal sampling rate:

- REC0058: 20,480 Hz
- REC0131: 20,480 Hz

Sampling rate after the first order correction:

- REC0058: 20,480.026214158832 Hz
- REC0131: 20,480.011221929413 Hz

The plot provided in Figure 2.12 shows these two signals in the time zone of the triggering point.

The plot provided in Figure 2.13 shows the transient event at the end of a recording without a sampling rate correction. It shows the time difference between two signals is about as large as 36 sampling points.

The time difference shown in Figure 2.12 is caused by using the nominal sampling rate, which is inaccurate.

The plot in Figure 2.14 shows the same transient event at the end of a recording while the first order correction is applied to the sampling rate of both signals.

The two signals are lined up beautifully on the horizontal axis.

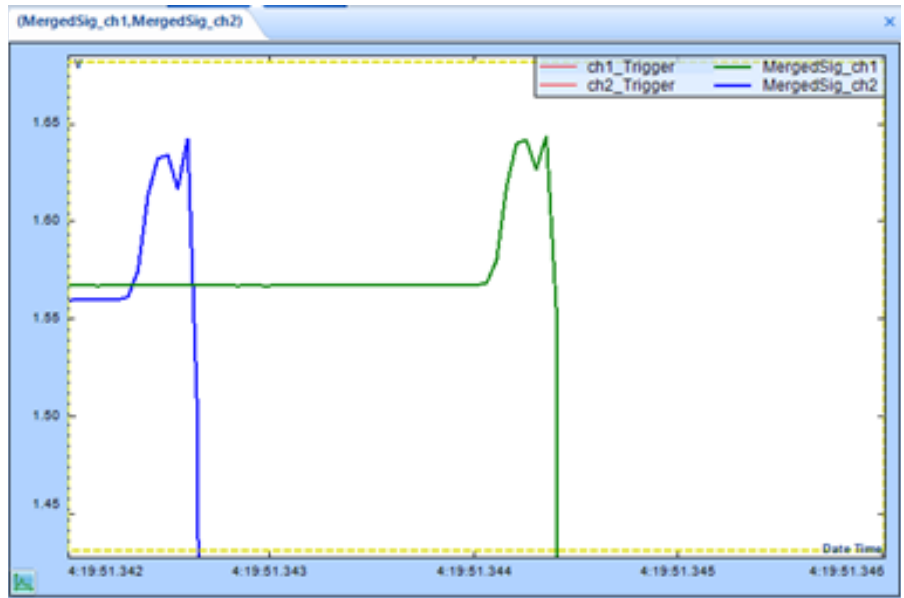


Figure 2.13

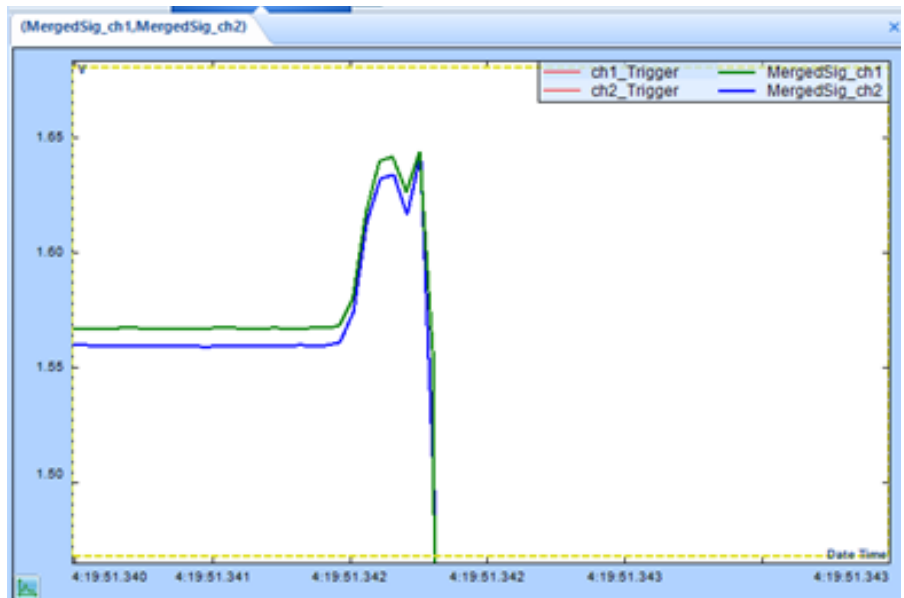


Figure 2.14

Spectral Analysis based on Time Stamped Signals

Summary

With the recorded signals from multiple GRS units, the processing algorithm can be summarized in the following steps:

1. First, the user will need to enter typical measurement parameters such as data window, block size, overlapping ratio, average number, etc. into the Crystal Instruments Post Analyzer software.
2. The calculation of cross spectra is the same as before, except:
 - a. The starting point of the second channel, which is derived from the overlapping ratio and block size, will be calculated with a new algorithm described below.
 - b. After the cross spectrum is calculated, a phase adjustment will be made.
3. Auto-power spectra of excitation and response channels shall be calculated as before.
4. With the auto-power and cross-power spectra, the FRF can be derived.

Calculate the starting point of a time signal block

If the digital data acquired from both channels is completely synchronized, which is the case when they are from one integrated dynamic signal analyzer, the blocks from both channels can be calculated based on the overlapping ratio and block size: <https://www.crystalinstruments.com/dynamic-signal-analysis-basics>

Overlap Processing

To increase the speed of spectral calculation, overlap processing can be used to reduce the measurement time.

The diagram provided in Figure 3.1 shows how the overlap is realized.

As shown in this diagram, when a frame of new data is acquired after passing the Acquisition Mode control, only a portion of the new data will be used. Overlap calculation will speed up the calculation with the same target average number. The percentage of overlap is called overlap ratio. 25% overlap means 25% of the old data will be used for each spectral processing. 0% overlap means no old data will be reused.

When two signals are from different GRS units—and because the sampling rate of two systems is slightly different—there will be a different number of sample points for the same duration of acquisition.

Therefore, we cannot simply use the overlapping ratio to calculate the starting point of the block for each

channel. Instead, we will calculate the starting points of two channels with the following algorithm:

1. Use the same algorithm based on block size and the overlapping ratio to calculate the starting point of the first channel (X channel).
2. To calculate the starting point of the block of the second channel (Y channel), using time stamps, we look for the point that is nearest to the starting point of the first input channel in absolute time. Assume the time difference between these two points is dt .

With the starting points of both channels known, calculate the instantaneous cross spectrum between two channels, $S_{yx} = S_x * S_y * T$, using the same formula:

<https://www.crystalinstruments.com/dynamic-signal-analysis-basics>

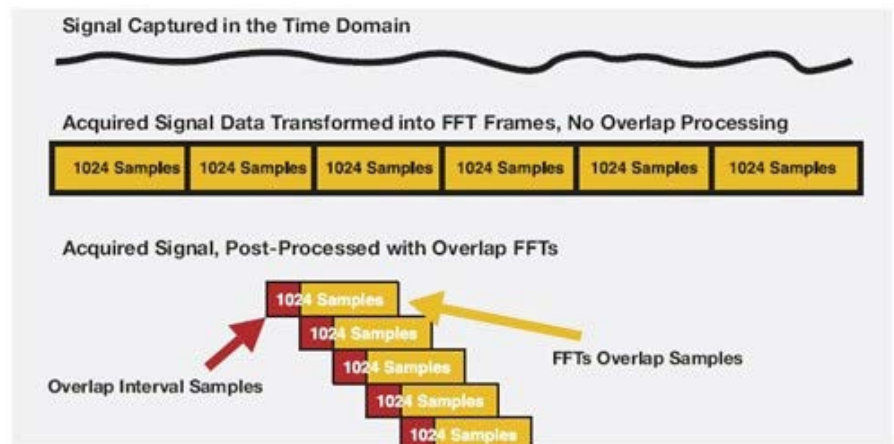


Figure 3.1 - Illustration of overlap processing

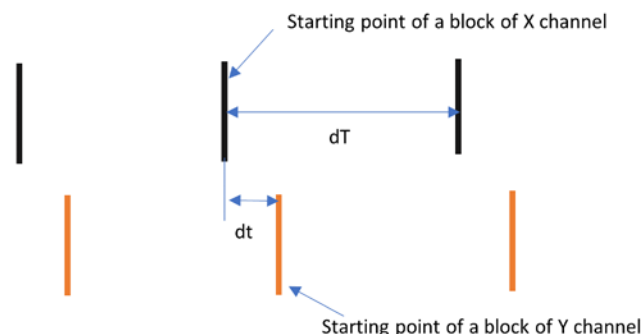


Figure 3.2

Cross Spectrum Computation

Cross spectrum or cross power spectrum density is a frequency spectrum quantity computed using two signals, usually the excitation and response of a dynamic system. Cross spectrum is not commonly used on its own. Most often it is used to compute the frequency response function (FRF), transmissibility, or cross correlation function.

Follow the steps below to compute the cross-power spectral density G_{yx} between channel x and channel y:

Step 1 - compute the Discrete Fourier Transform of input signal $x(k)$ and response signal $y(k)$:

$$S_x = \sum_{n=0}^{N-1} x(k) w(k) e^{-j2\pi kn/N}$$

$$S_y = \sum_{n=0}^{N-1} y(k) w(k) e^{-j2\pi kn/N}$$

Step 2 - compute the instantaneous cross power spectral density:

$$S_{yx} = S_x * S_y * T$$

Step 3 - average the M frames of S_{yx} to obtain the averaged cross power spectrum: G_{yx}

$$G_{yx} = \text{average}(S_{yx})$$

Note: there are many ways to average the spectrum, including linear average, moving linear average, exponential average etc. Regardless of the method used, the same method should be applied to both auto and cross spectra in order to compute the transfer and coherence functions.

Step 4 - Compute the energy correction and double the value for the single-sided spectra.

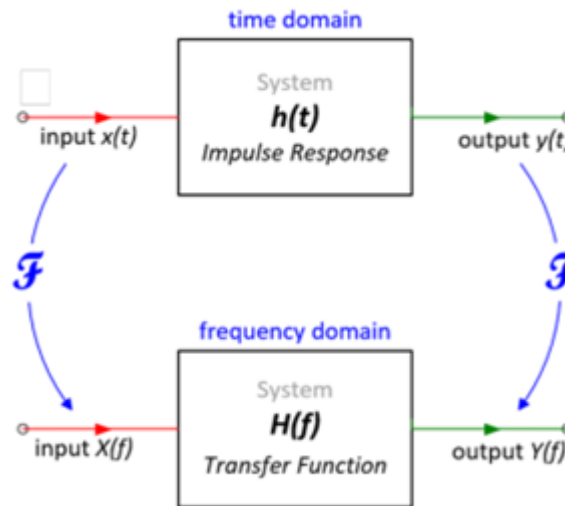


Figure 4.1

$$G_{yx} = 2 G_{yx} / \text{EnergyCorrection}$$

After the cross spectrum is calculated, the system transfer function (H1 type is shown here) can be derived by:

$$H_{yx} = G_{yx} / G_{xx}$$

Where G_{xx} is the averaged auto-power spectrum.

Apply the phase correction to cross spectrum $S_{yx}[n]$

With time stamped signals, the cross spectrum and transfer function calculation process will remain the same except that we will add a phase correction process after the instantaneous cross spectrum is calculated.

S_{yx} is called instantaneous cross spectrum because it has no average effect in it. It is an array with N pairs of complex numbers and its nth element is $S_{yx}[n]$. After S_{yx} is calculated, we will conduct a phase correction to each point of the signal $S_{yx}[n]$.

We know the data from two input channels has a known time delay of dt. The phase correction formula will apply the following phase change to each element:

$$\text{Phase change}[n] = 2\pi * (dt/dT) * n/N$$

Where:

- dt is the time delay between the starting point of the block of Y channel vs. X channel
- dT is the nominal sampling interval. For example, for a sampling frequency of 102.4 kHz the dT is 9.765625 μ s.
- n is the index number of the element in the array
- N is the total size of the array

The formula above simply indicates that if the time delay is 1 sample point long, the phase shift between two channels will be 2π , or 360 degrees, when frequency index is equal to the sampling frequency. Figure 4.2 demonstrates how it can be described in the frequency domain.

Now, the question is how to make a phase correction to a complex number.

If a complex number is expressed in an amplitude and its phase, we can simply add the *Phase change[n]* value to the phase portion θ .

But if the complex number is expressed in real and imaginary parts, you will have to transform them into amplitude/phase, add phase, then transform it back to real/imaginary parts.

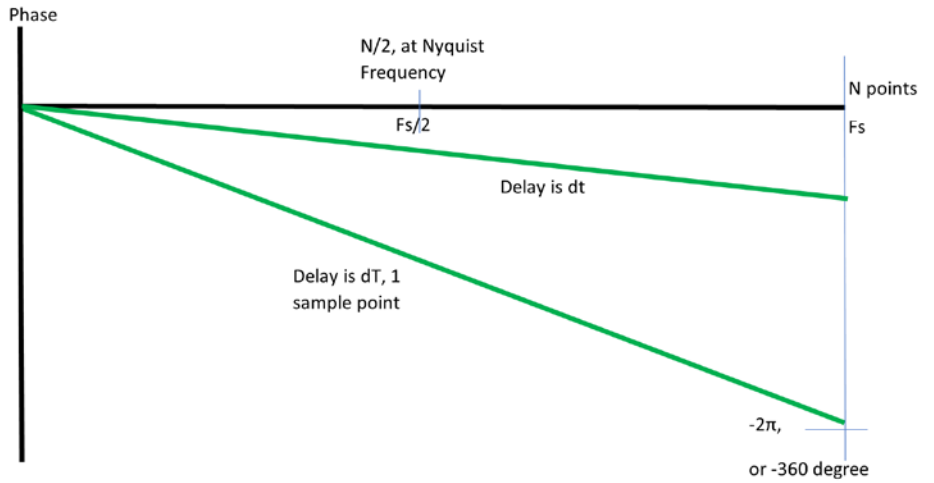


Figure 4.2

Conduct the average and remaining spectral calculations

After the phase correction is made to the instantaneous spectrum S_{yx} , it will be averaged into G_{yx} , the averaged cross spectrum, using either a linear or exponential averaging algorithm. Then all other spectra, FRF, Coherence etc., can be calculated as usual.

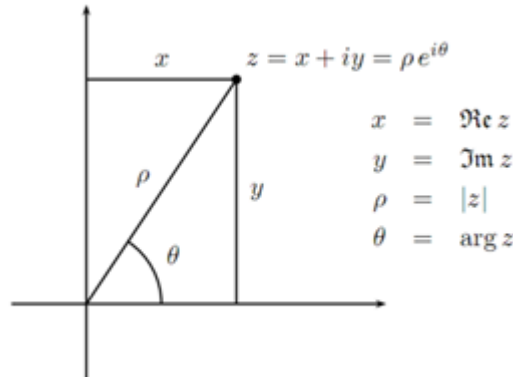


Figure 4.3

Data processing diagrams

The GRS data acquisition systems can be physically separated by hundreds of feet or miles without electrical hardware connections. The systems will instead receive GPS signals from the satellites and certain commands from a remote computer through wireless communication methods.

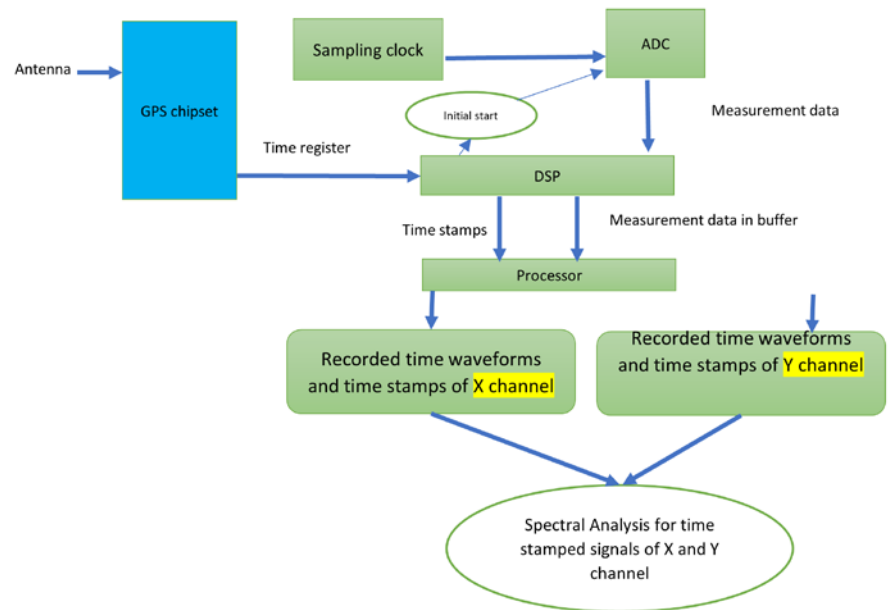


Figure 4.4

Figure 4.4 describes the general data flow using time stamped signals.

The processing block at the bottom, Spectral Analysis for time stamped signals, can be expanded to following diagram shown in Figure 4.5.

The interesting part of this technique is that while the misalignment of sampling clocks in the time domain is obviously caused by deviations of the ADC sampling clocks from two data acquisition units, the correction is not performed in the time domain. Instead, it is performed in the

frequency domain of cross spectrum. In fact, we believe that there is no good algorithm to correct the arbitrary time delay between X and Y channels with two separate ADC clocks.

Longest GPS antenna cable

In certain applications the GRS hardware is not accessible to the signals sent from the GPS satellites. A longer antenna cable is needed.

CI has tried several antenna models with different lengths. The longest antenna meeting the requirement that has been tested is 150 feet long. This test indicated that the GRS can be positioned in a location that has no direct exposure to the GPS signals as long as the terminal of its GPS antenna is accessible to the GPS signals.

Spectral Analysis for time stamped signals of X and Y

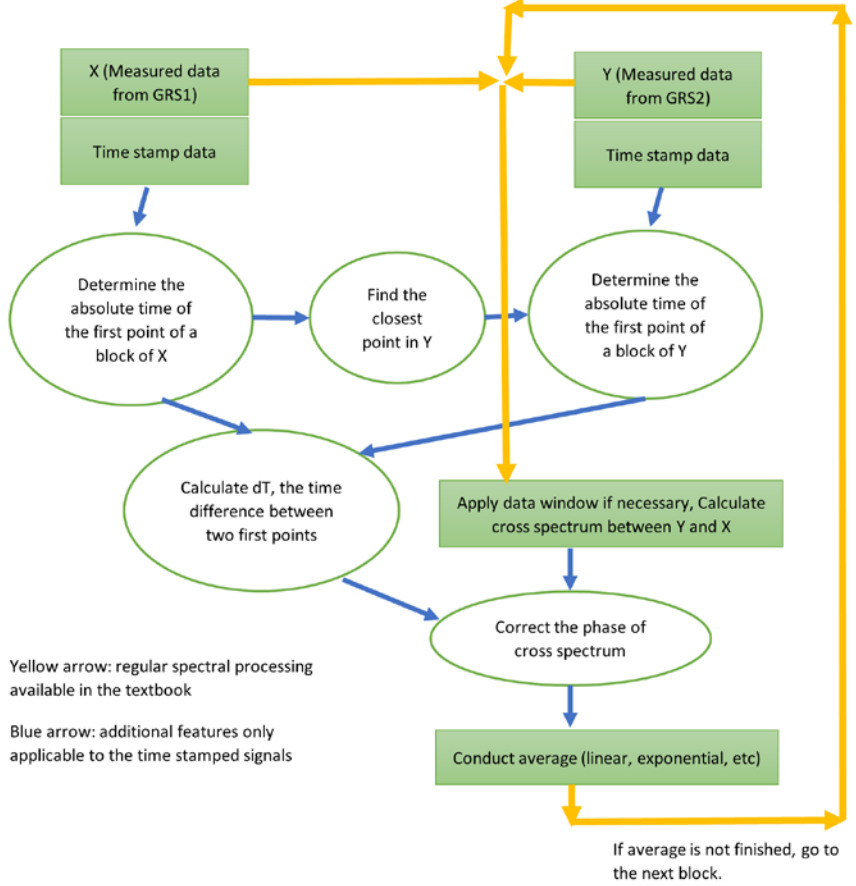


Figure 4.5



Figure 5.1

Examples of Applications

Bare-wire testing

To validate the effectiveness of the algorithm above, we established the following test. We took two GRS units, each with the GPS time stamping function installed. We fed white noise into one channel of GRS unit 1 and one channel of GRS unit 2. The same white noise source can be split into two using a simple BNC splitter.

We set both GRS units with a nominal sampling rate of 102.4 kHz and started to acquire the data. All measurement data can be recorded into the SD card. By design the total recording duration can be hundreds of hours, but we only recorded a few minutes for this particular test.

After the recording, we transmitted the measured data files together with their time stamp files to a PC using GRS-Host software. The GRS-Host software can directly read the data files on the SD card or read the data files from each GRS when Ethernet is connected.

After the data files are located on the PC, we opened the Post Analyzer software (PA). PA can merge multiple data files into one and then analyze it.

After the measured data files are merged based on time stamping information, they can be analyzed using the technique mentioned above. In order to demonstrate the effectiveness of this method, we set the average number to 1, which means no averaging over multiple blocks of data. If we apply averaging, then the phase measurement will not be valid.

The screenshot in Figure 6.1 shows a typical test result. The top trace shows the original time data from both GRS units. The middle trace

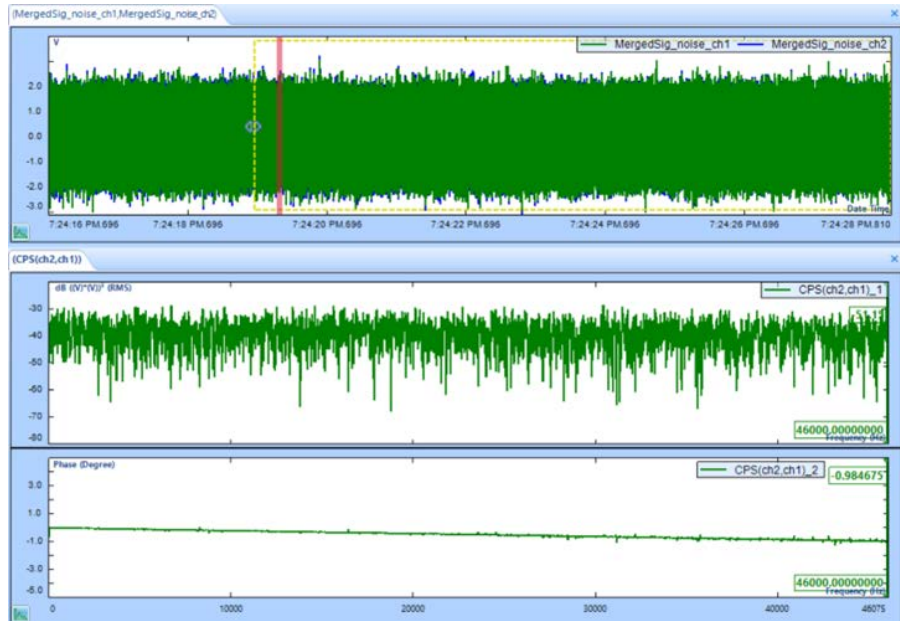


Figure 6.1

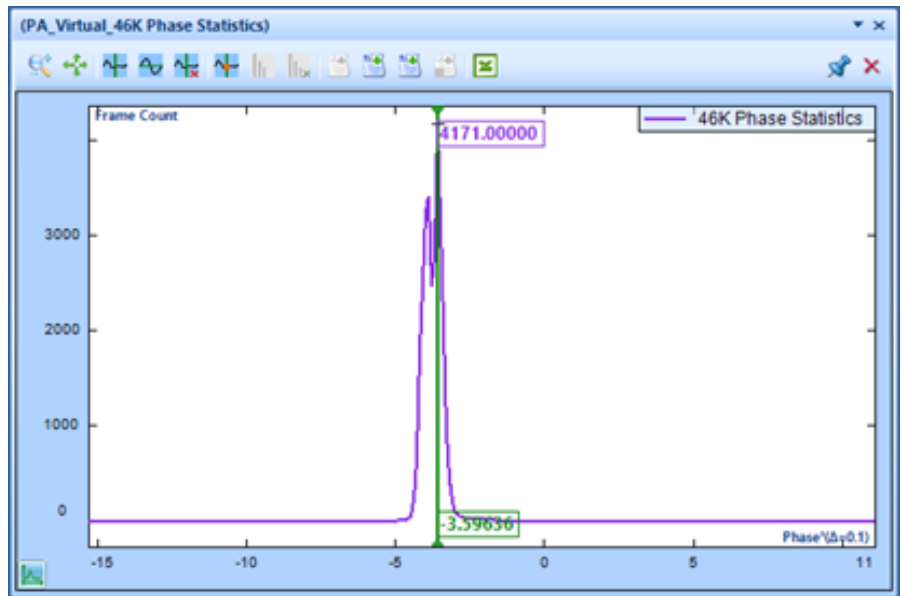


Figure 6.2

shows the amplitude of the cross-spectrum of this particular block. The bottom one, which is what we really care about, shows the phase spectrum of this block from 0 Hz to 46 kHz. A cursor located at 46 kHz has a value at -0.984675 degree. This is the phase measurement at this frequency.

In order to establish a statistical confidence of the phase deviation, we calculated thousands of blocks cross-spectra and **stored all of their phase values at 46 kHz**. After these values

are stored, a histogram like the one in Figure 6.2 can be drawn.

In this histogram plot, the vertical axis indicates the block counts, i.e., the number of phase measurements at 46 kHz. The horizontal axis is the phase in degrees. By measuring thousands of frames of cross-spectra, we are able to determine the statistical distribution of the phase error. This is how we can claim the phase measurement accuracy listed in the following table:

Phase match between two GRS units:	±5°(degree) at 40 kHz
	±2.5°(degree) at 20 kHz
	±0.5°(degree) at 2 kHz

We used multiple GRS units for testing, and they all showed similar characteristics.

The phase value can be converted directly to the time delay of that frame between two channels. The formula is:

$$dT = 1/46 \text{ kHz} * \text{phase}/360$$

For example, if the phase is 5 degrees, then the time delay between two channels of that frame is:

$$dT = 1/46 \text{ kHz} * 5.0/360 = 3e-7 \text{ (second)} = 300 \text{ ns}$$

Calculating the auto-spectrum is relatively easy because it only needs to deal with the data from a single channel. Once auto-spectra and cross-spectrum are calculated, we can compute the Frequency Response Function and the coherence function as shown above.

Testing results using an electrical simulation box

Crystal Instruments developed an electrical simulation box (Figure 6.3) to simulate the resonance of a structure. It is a perfect tool to validate the effectiveness of the time stamped signal processing algorithm.

By connecting the simulation box to a signal analyzer with random excitation (Figure 6.4), we are able to measure the transfer function (FRF).

In order to validate the effectiveness of Time Stamped Signal Processing described in this document, we can connect two GRS units to acquire the data (Figure 6.5).



Figure 6.3

Signal Analyzer

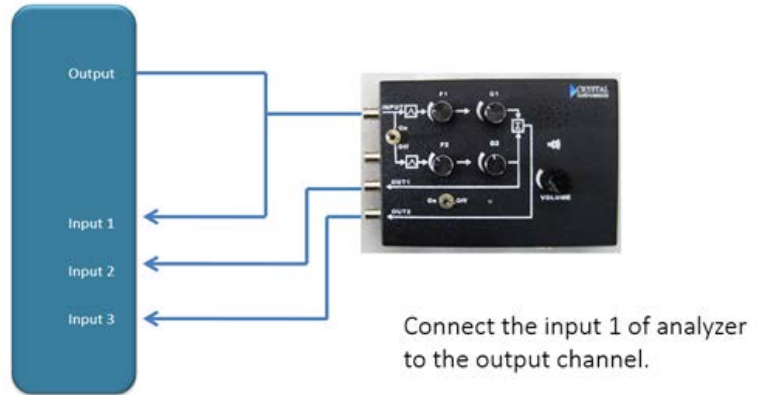


Figure 6.4

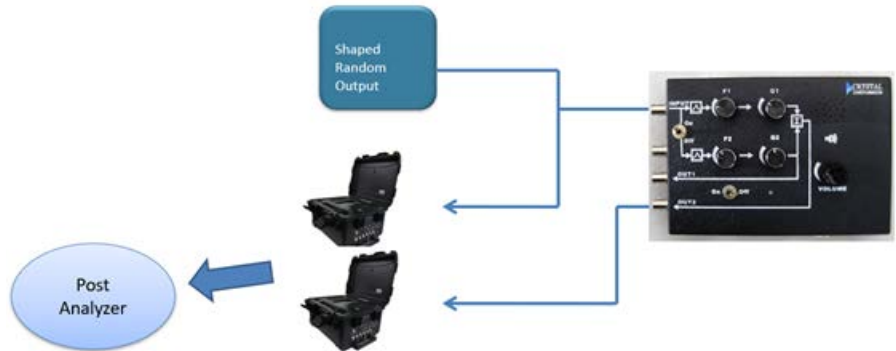


Figure 6.5

With two different methods of measurement, we can calculate the FRF and make the comparison (Figure 6.6).

Blue: FRF measured using single Spider instrument

Red: FRF measured with two GRS units, using Time Stamped Signal Processing (the red signals in the plot are nearly invisible because they are covered by the blue ones).

The two measurement signals are nearly identical. This comparison further proves that the Time Stamped Signal Processing can generate the same results as using one single instrument.

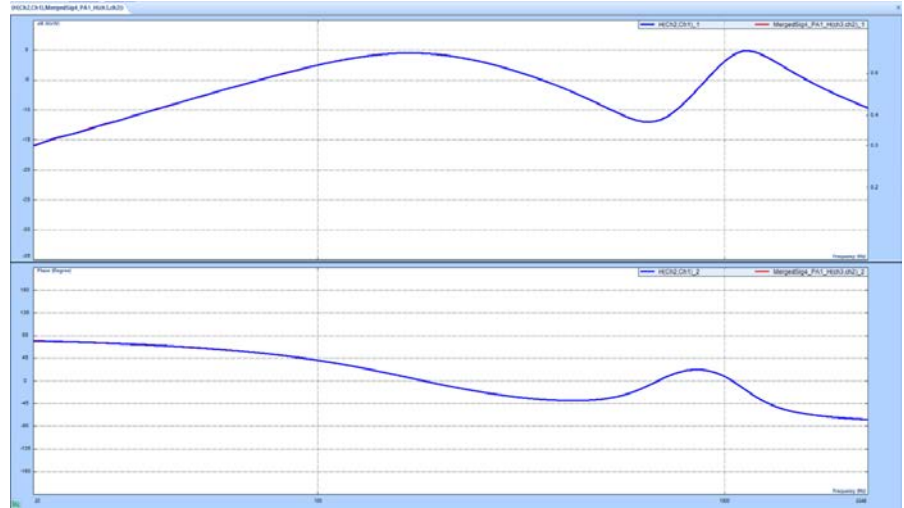


Figure 6.6

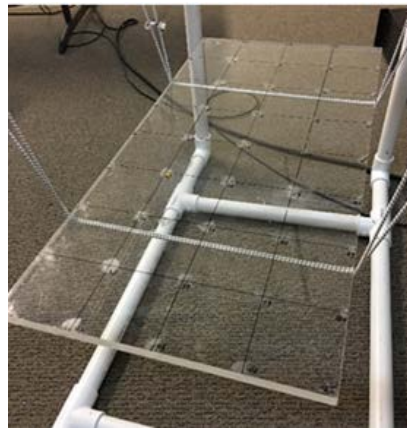


Figure 6.7

Modal Testing

Objective

A Frequency Response Function (FRF) is first calculated using a conventional instrument, the Spider system. Its testing results are compared to those obtained using two separate GRS systems, one system measures excitation and one system measures response. While the test is conducted on a small structure, the method is applicable to any large mechanical structures such as vessel ships, bridges, or buildings.

Test Results

FRF Measurement using Hammer Impact Modal Test on a Plexiglass Plate

1. A hammer impact modal test is carried out on a plexiglass plate for one measurement point. (Figure 6.7)
2. The raw time stream data for the excitation is recorded on one GRS system. The raw time stream data for the response is recorded on the second GRS system. This data is later transferred to PA software for post-processing and calculating the FRF. (Figure 6.8)

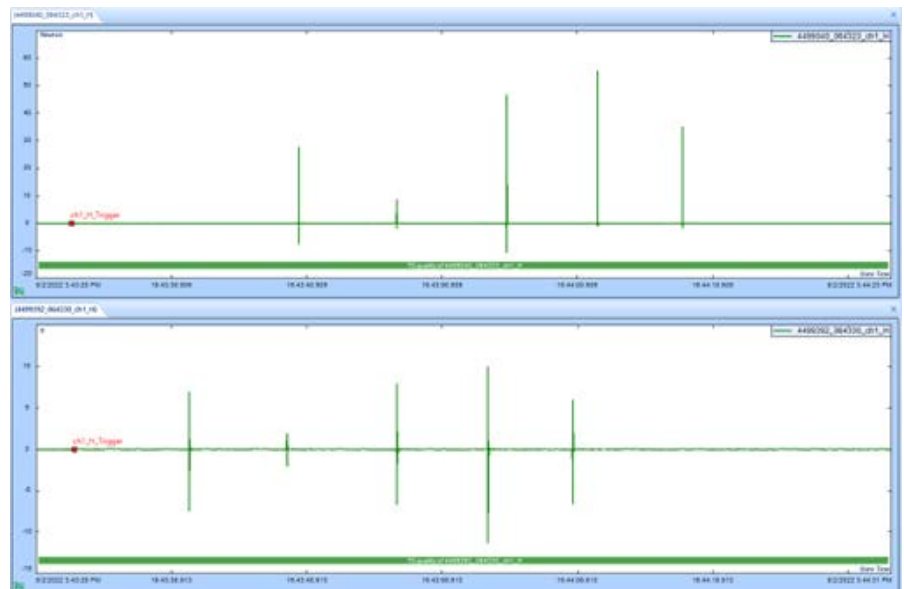


Figure 6.8

3. The Spider system also measures the excitation from the hammer impulse and the response from the accelerometer response. The FRF is calculated using EDM Modal software. (Figure 6.9)

4. The configuration settings are appropriately set and are kept consistent across the different hardware systems and EDM software. (Figure 6.10)

5. The FRF plots are overlaid to compare the data from the GRS and Spider system. The green curve is generated using the Spider-80X's acquired data and the blue curve is generated using the data acquired from the two GRS systems. (Figure 6.11)

This test indicates that two separate GRS units can measure FRF and other modal parameters to obtain identical results as using conventional instruments, while the latter has to be wired and is limited by the distance between their measurement points.

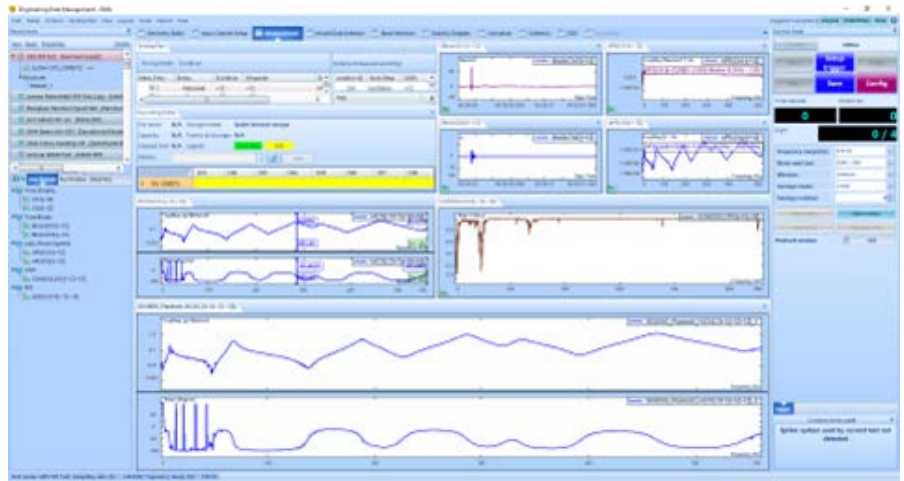


Figure 6.9

Frequency range(Hz):	576 Hz
Block size/Line:	2048 / 900
Window:	Uniform
Average mode:	Linear
Average number:	4

Figure 6.10

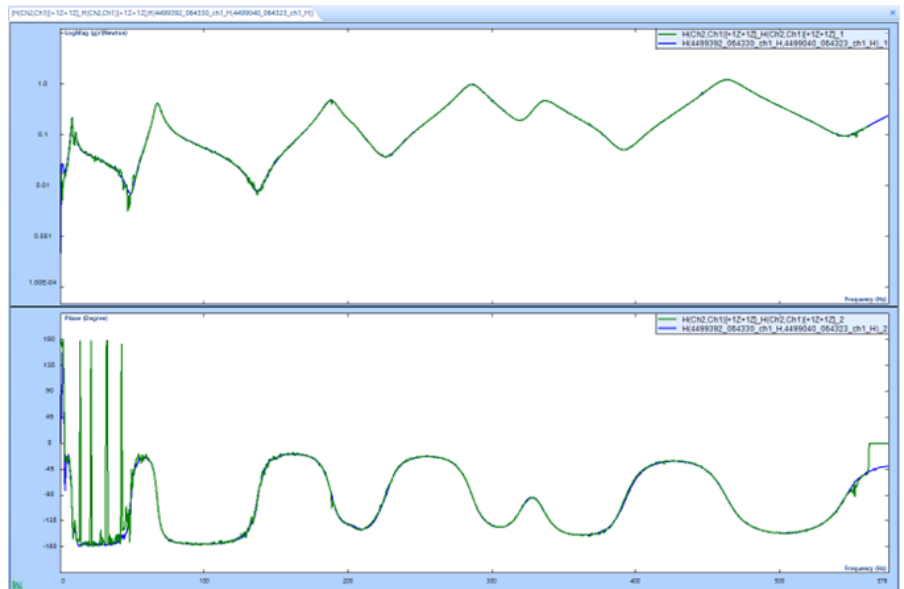


Figure 6.11

Sound source identification (locating the coordinates of a gunshot)

If the earth is flat, we would only require three GRS units to measure the sound of a gunshot and thereby determine the location of its shot. If the assumption is that the earth is not flat, four GRS units will be required to determine the location of the sound source.

Known: locations of A1, A2, A3
(GPS on GRS can provide very accurate location information)

Known: distance of B2-B1, B3-B2, B3-B1 (the difference between any pair). These can be measured using correlation functions of any pair of sound pulses. These distances can be estimated by using the following method:

1. Use any pair of time domain signals with time stamps to calculate the cross-correlation.
2. Using cross-correlation, detect the peak location.
3. Based on the peak location, determine the time delay between two signals.
4. Based on the time delay and speed of sound, determine the distance.

Then the position of the sound origin can be calculated.

If the three GRS units and the sound source are not in the same plane, we will require a fourth GRS to determine its location. A simple software is developed based on the cross-correlation method to calculate the delay between each pair of GRS. With these three time delays and the assumed speed of sound, the sound source location can be calculated.

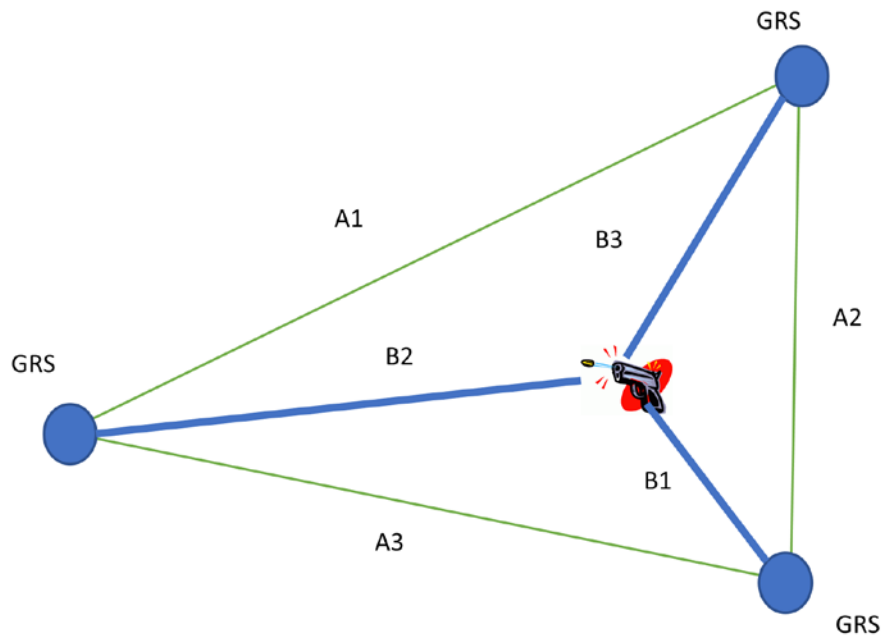


Figure 6.12

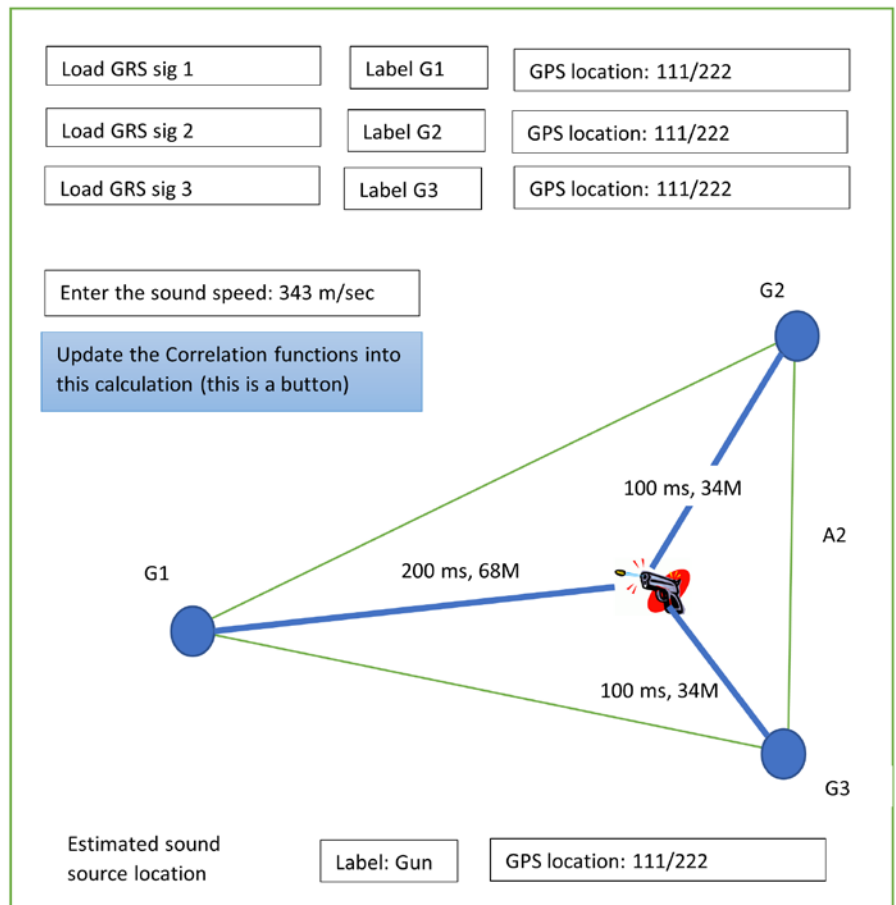


Figure 6.13

In the summer of 2022, CI conducted a successful test at the Los Altos Rod and Gun Range in Los Gatos, California. The results were excellent. (Figure 6.13)

Figure 6.14 shows a screen shot of three signals over-plotted on the same window.

Using the recorded signals, we used cross-correlation methods to calculate the gunshot source location. It can be accurately displayed on the map as shown in Figure 6.15.

For details regarding this test, please contact Crystal Instruments to request more information.

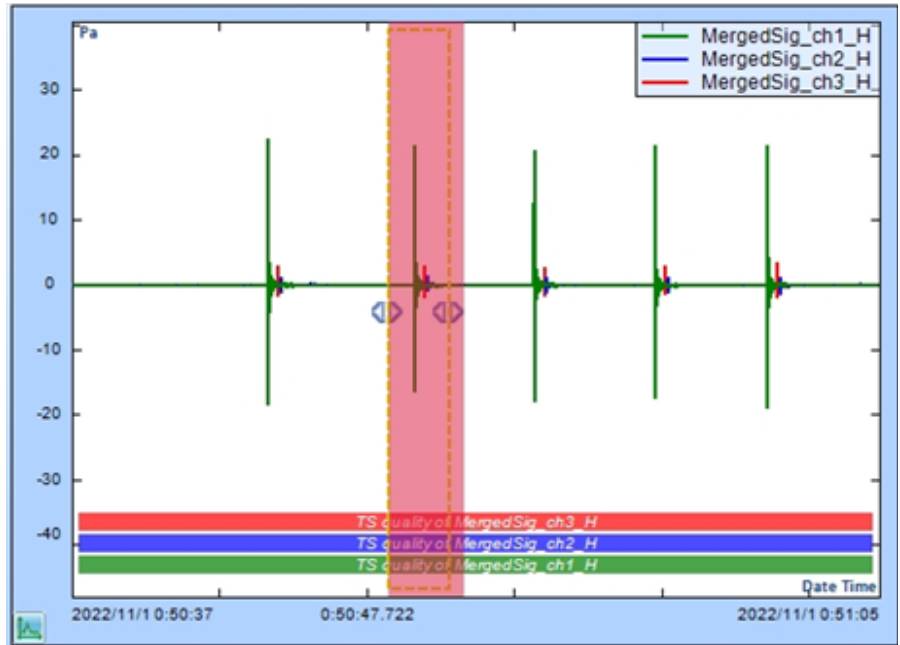


Figure 6.14

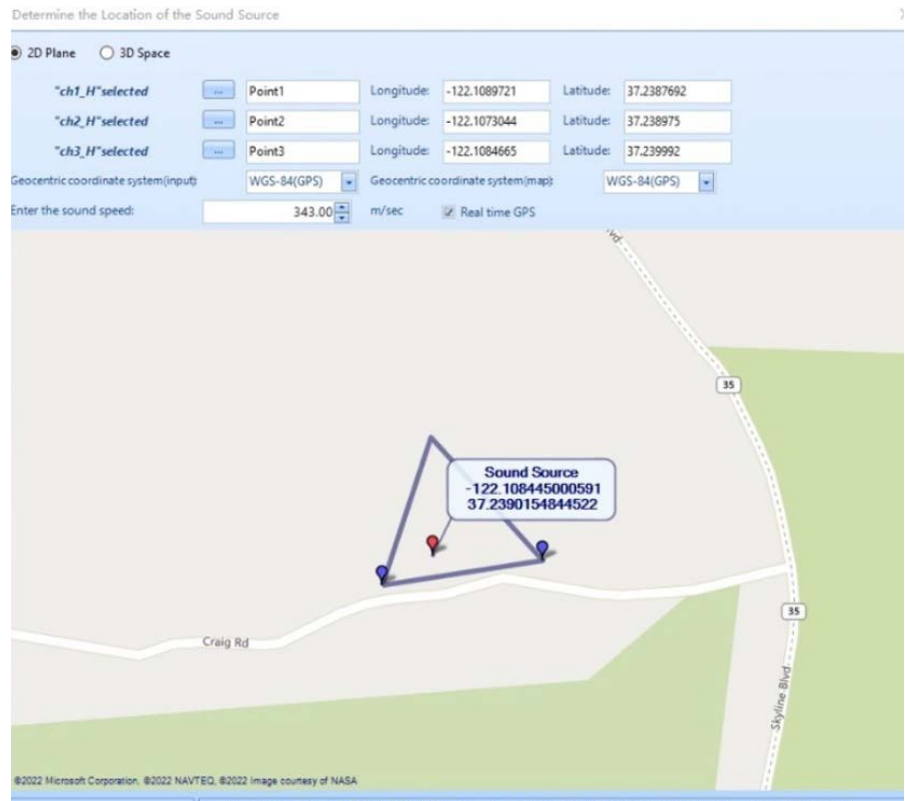


Figure 6.15

Crystal Instruments Corporation
2090 Duane Avenue
Santa Clara, CA 95054

Crystal Instruments Testing Lab
15661 Producer Lane, STE H
Huntington Beach, CA 92649

Crystal Instruments Testing Lab
1548A Roger Dale Carter Boulevard
Kannapolis, NC 28081

Phone: +1 (408) 986-8880
Fax: +1 (408) 834-7818
www.crystalinstruments.com