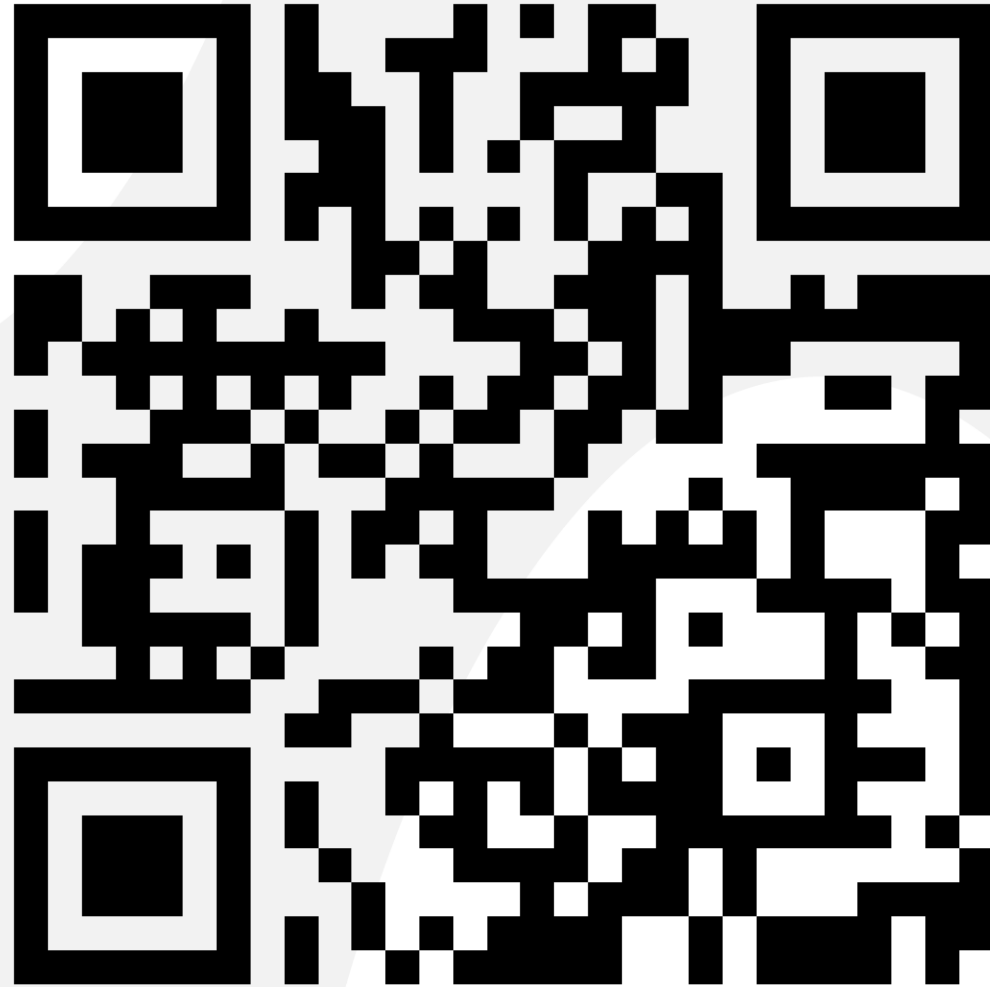


# Welcome to the Keysight Technologies Students Workshop

08 November 2022  
Universität Stuttgart

If you want to get a **certificate of the workshop completion**, please register here:





## Workshop

# Fundamentals of Arbitrary Waveform Generation (AWG)

Vitaly Morarenko

Solutions Engineering  
Keysight Technologies



# Agenda

## AWG Workshop

Arbitrary Waveform Generation Fundamentals

Frequency Response Correction

In-System Calibration

Conclusion & Summary

# Agenda

## AWG Workshop

Arbitrary Waveform Generation Fundamentals

Frequency Response Correction

In-System Calibration

Conclusion & Summary

# History of Arbitrary Waveform Generators

From analog to digital signal generation

Sep 1950

Signal Generator with internal and external modulation capabilities



HP 618

April 1988

Arbitrary Waveform Synthesizer with digital architecture and analog output, 50 MHz



HP 8770A back view

Sep 2018

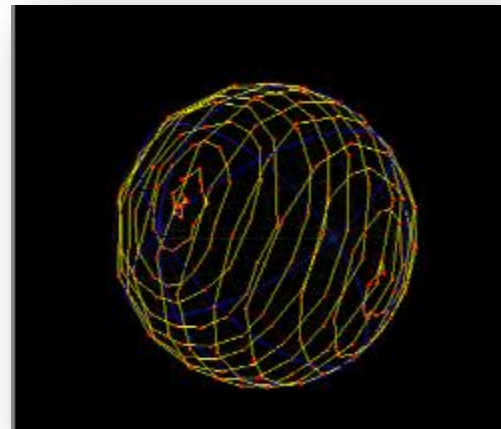
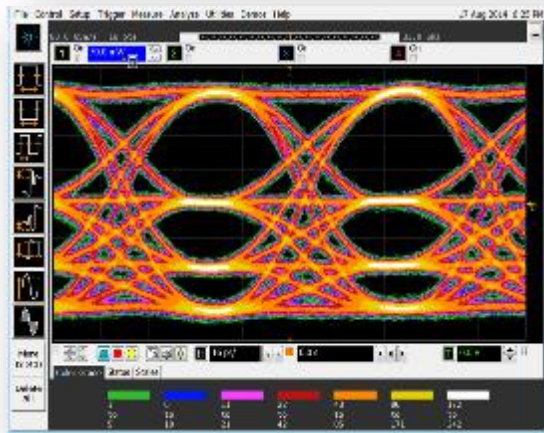
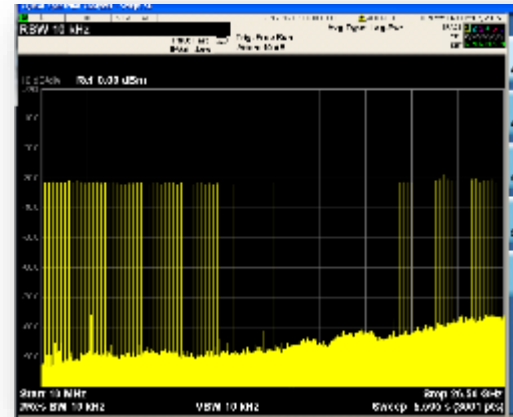
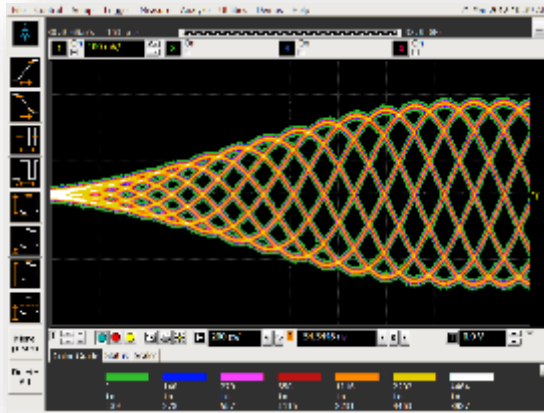
Arbitrary Waveform Generator with 120 GSa/s 45 GHz analog bandwidth



Keysight M8194A

# An ARBITRARY Waveform Generator

The most versatile signal scenario generator possible



It demonstrates the flexibility you can have with an AWG. Whatever you can described mathematically you can generate.

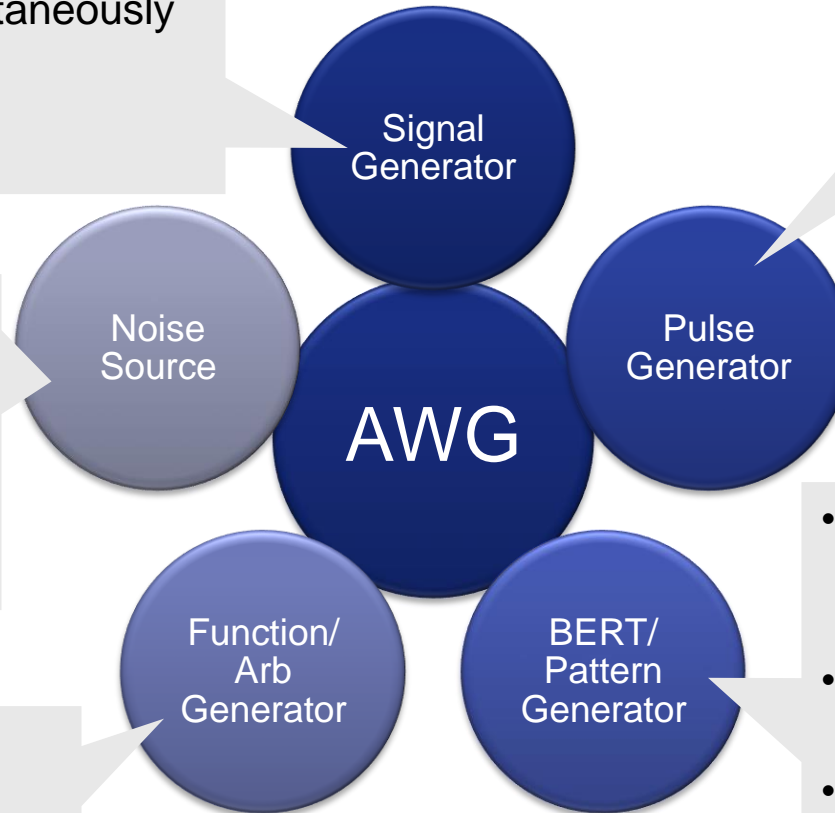
...you are only limited by your imagination!

# AWG in Comparison to Other Instruments

- AWG offers wider modulation BW
- Generate multiple carriers simultaneously
- Fast hopping
- BUT: less dynamic range

- AWG offers more flexibility, e.g. custom spectral “shapes”, notches, narrowband noise, etc.
- BUT: not as “random” as true noise sources

- AWG has typically a superset of functionality
- BUT is typically more expensive (for same bandwidth)



- AWG offers more flexibility, e.g. different pulse shapes or adding pre-distortion
- BUT new waveform required for every pulse parameter change

- AWG offers more flexibility, e.g. variable rise times, multi-level signals, pre-distortion
- BUT can not do true RJ and limited pattern length
- No error detector included



# Find the AWG That's Right For You

Push your design to the limit

The M8100 Series arbitrary waveform generators (AWGs) offer a level of versatility that enables you to set up complex real-world signals — whether you need precise signals to characterize the performance of a design or need to stress a device to its limits. From low-observable radar to high-density communications, testing is more realistic with our precision arbitrary waveform generators.



**M8190A**  
12 GSa/s



**M8195A**  
65 GSa/s



**M8196A**  
92 GSa/s



**M8194A**  
120 GSa/s

# Keysight M8199A, 256 GSa/s, up to 70 GHz The World's Highest Performing AWG

Push your design to the limit



In order to create next generation technology, advanced research engineers require a new level of stimulus performance. Whether testing the discrete components of a coherent optical system or experimenting with terabit transmission, you need the highest speed, bandwidth, precision, and flexibility to meet the challenges of industry-leading applications. The unmatched capabilities of the M8199A enable you to take your designs to a new level.

# Keysight M8190A Arbitrary Waveform Generator

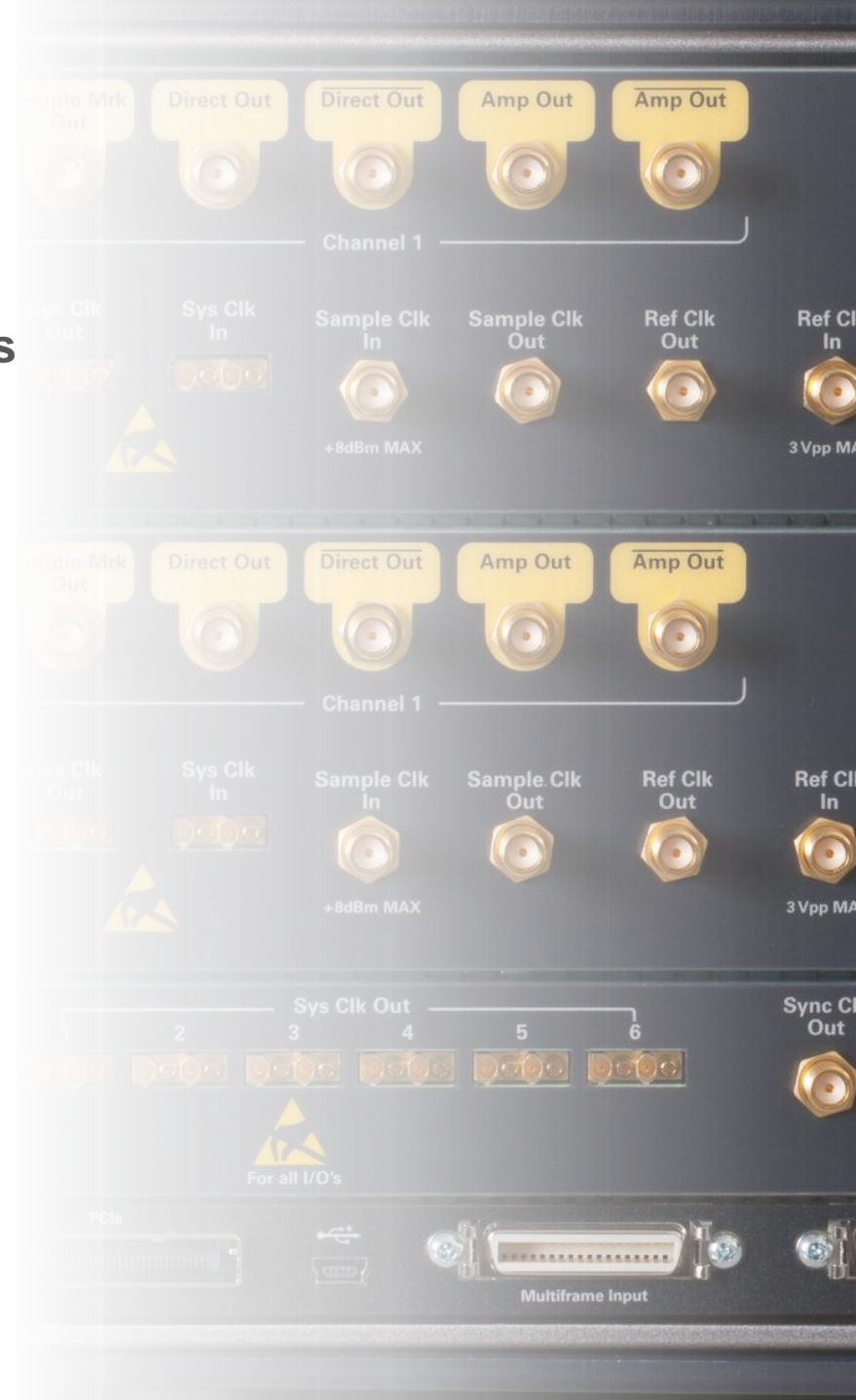
High-quality signal generation is the foundation of reliable and repeatable measurements. No matter the application, you must be confident that you are testing your device, not the signal source. The **high dynamic range and excellent vertical resolution** of the M8190A provides the accuracy and repeatability required to achieve the most reliable measurements possible.



# Key Benefits of the M8190A AWG

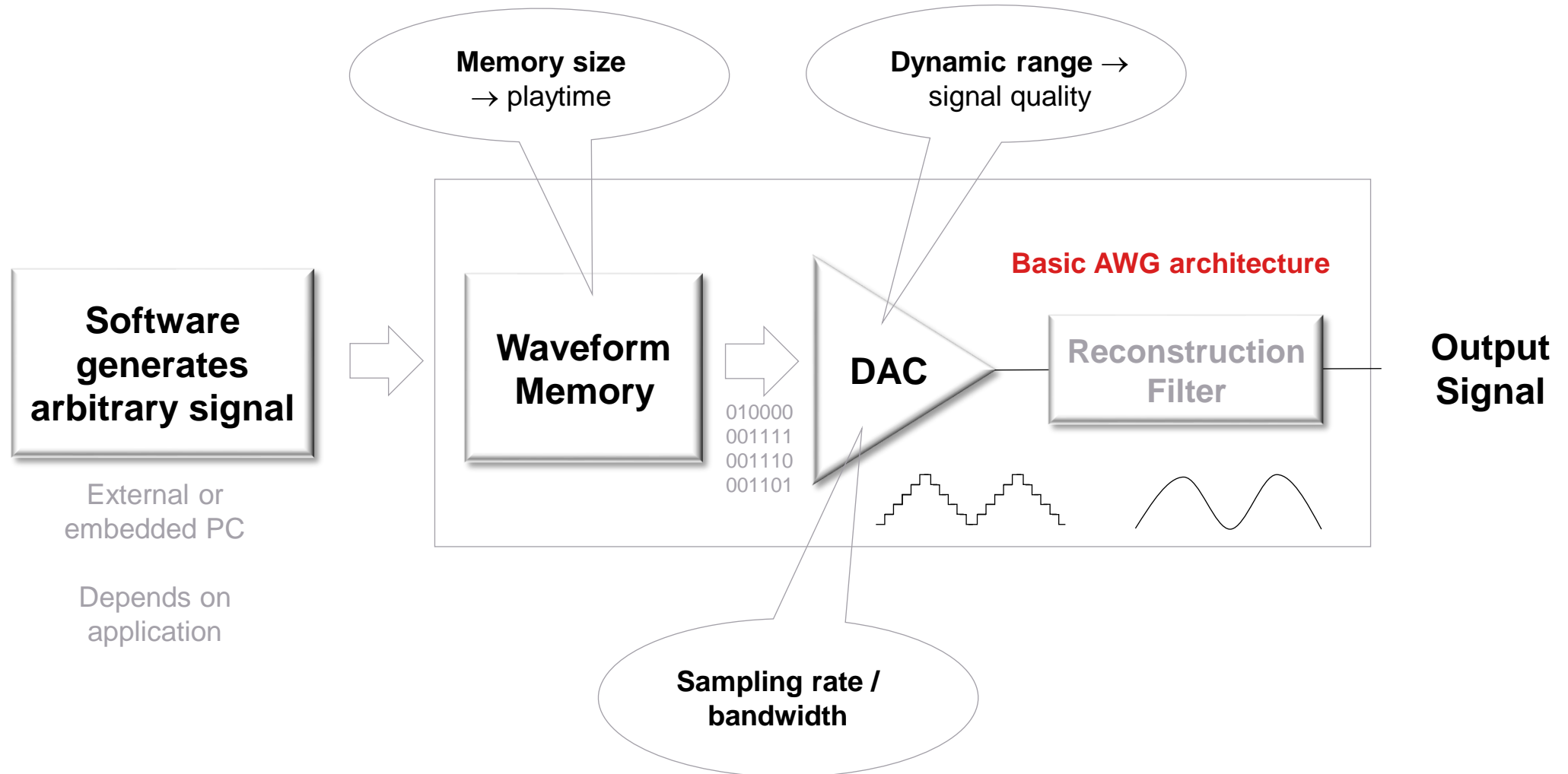
## Get the resolution you need

- Precision AWG with two DAC settings to **handle multiple applications**
  - **14-bit resolution** at 8 GSa/s – used to achieve highest vertical resolution
  - 12-bit resolution with up to 12 GSa/s – used to achieve highest sample rate
- Spurious-free-dynamic range (SFDR) up to 90 dBc ensures **tones clearly stand out**
- Analog bandwidth up to 5 GHz allows you to **mimic analog imperfections** with custom ISI
- 3 selectable output amplifier paths:
  - **Direct DAC** – optimized for best SFDR & high definition
  - **DC** – optimized for serial/data time domain applications
  - **AC** – optimized to generate high voltage, high bandwidth signals
- 1 or 2 channels per 2 slot module (sync up to 12 channels)



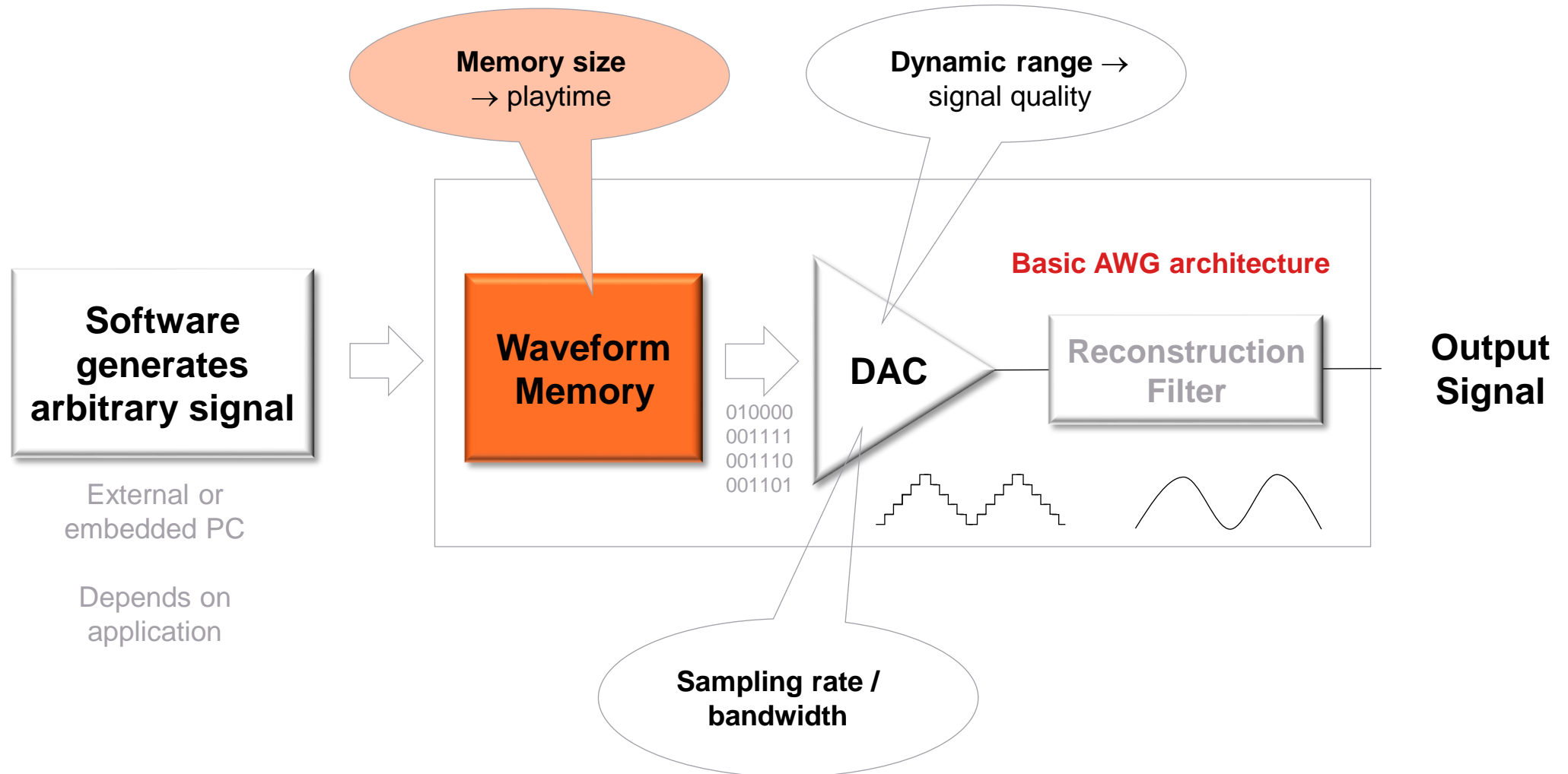
# Theory of AWG Operation

## Key Blocks and Specifications



# Theory of AWG Operation

## Key Blocks and Specifications: Memory



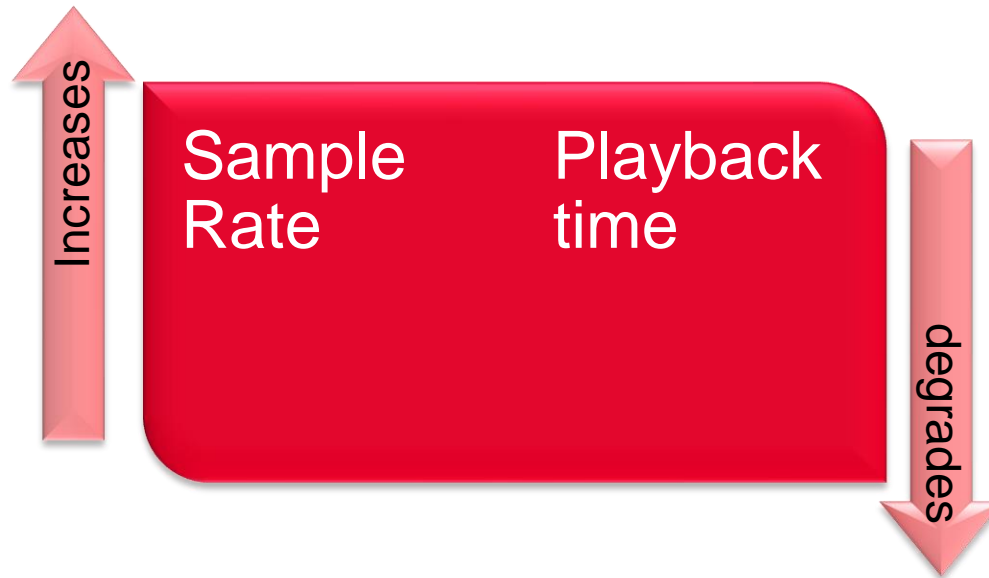
# AWG Basics: Memory

## Theory of AWG operation

**Memory ÷ sample rate = play time**

For example, 128 MSa ÷ 12 GSa/s = 10 ms

For example, 2 GSa ÷ 12 GSa/s = 166,67 ms



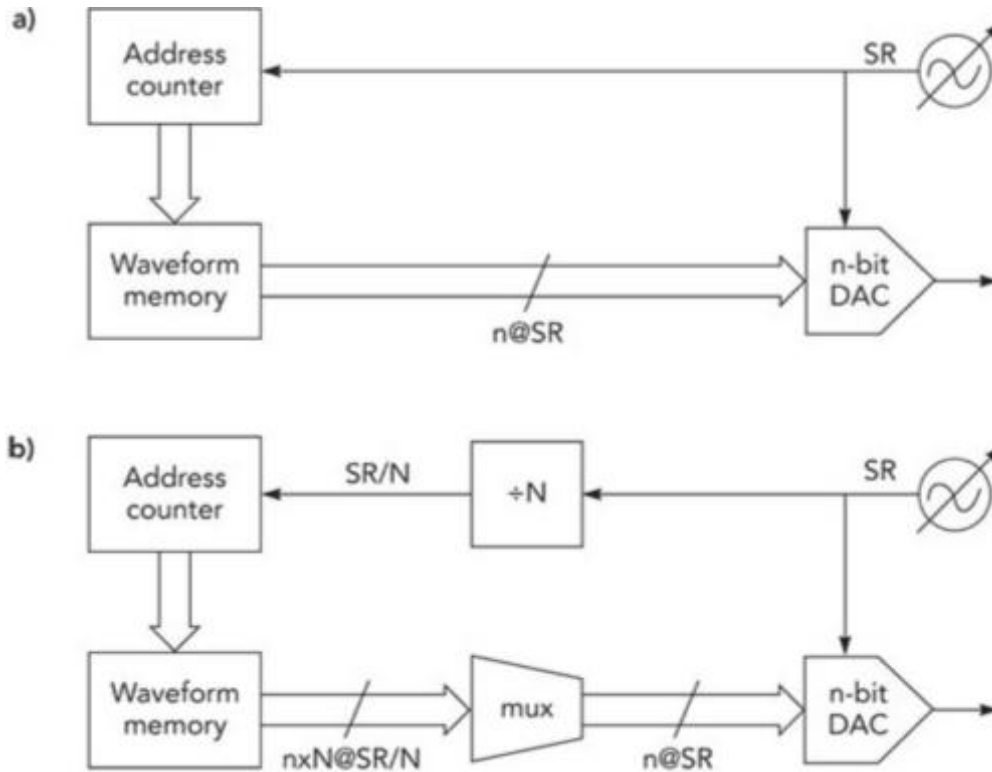
Waveform is defined in samples, which defines values in a given time interval.

The AWG outputs discrete output voltage levels

=> allows any different waveforms

# AWG Memory and Granularity

## Theory of AWG operation



Waveform memory access architectures are influenced by memory technology and access speed.

- a) Fast enough SRAM can be directly connected to the DAC
- b) If conversion speed is too high then it is possible to transfer more than one sample in a memory access cycle by widening the bus and using a multiplexer close to the DAC

### Granularity

The granularity reflects the number of sample that transfered in parallel from Memory to DAC

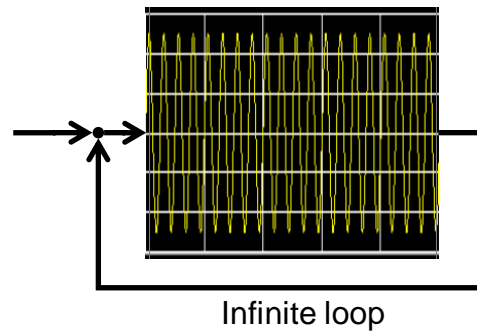


## Granularity and Min Length of Sample Memory of M8190A

Sample memory	
- Standard	128 MSa per channel
- Option 02G 12 bit mode	2048 MSa per channel
- Option 02G 14 bit mode	1536 MSa per channel
Option SEQ offers the enhanced sequencing functionality described below	
Minimum segment length	320 samples in 12 bit mode; 240 samples in 14 bit mode
Waveform granularity	64 samples in 12 bit mode; 48 samples in 14 bit mode

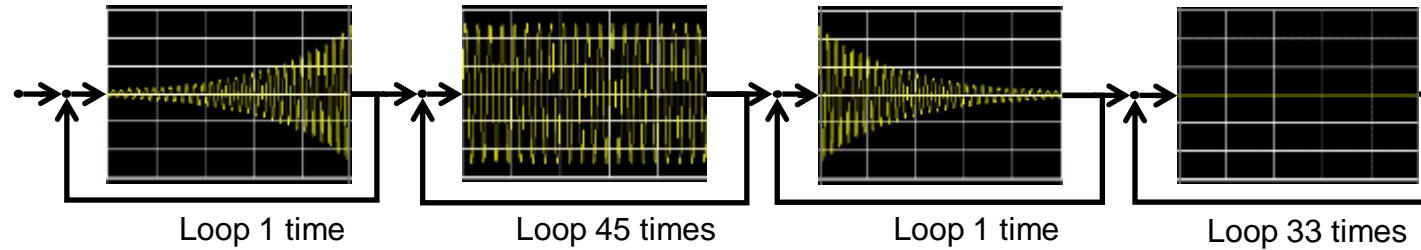
# AWG without Sequencing

- Only a **single** waveform segment is available
- Waveform segment can be up to full memory size

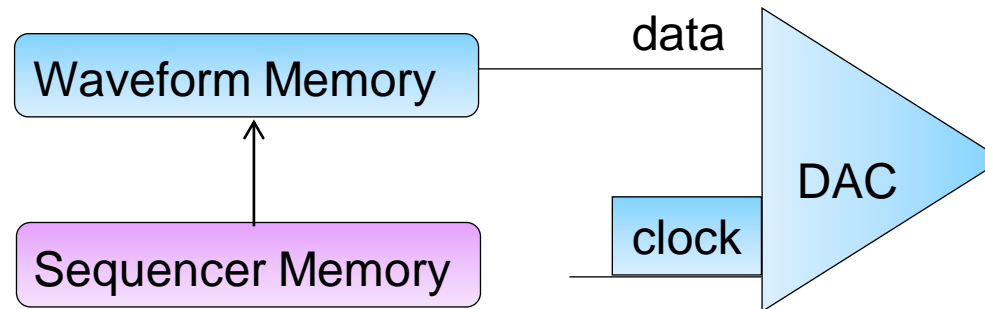


# AWG with Sequencing

How do we achieve longer playtimes?



Waveform segments are stored in memory

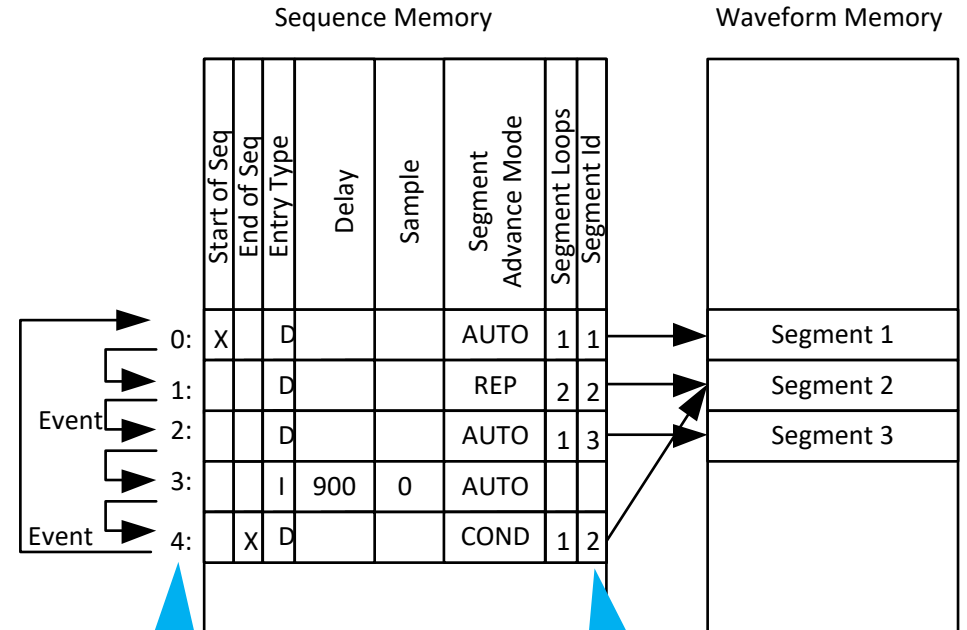


The Waveform Sequencer is where the waveform segments are arranged (sequenced) to create the desired waveform

**Memory ÷ sample rate ≠ playback time**

# Sequencing – Theory of Operation

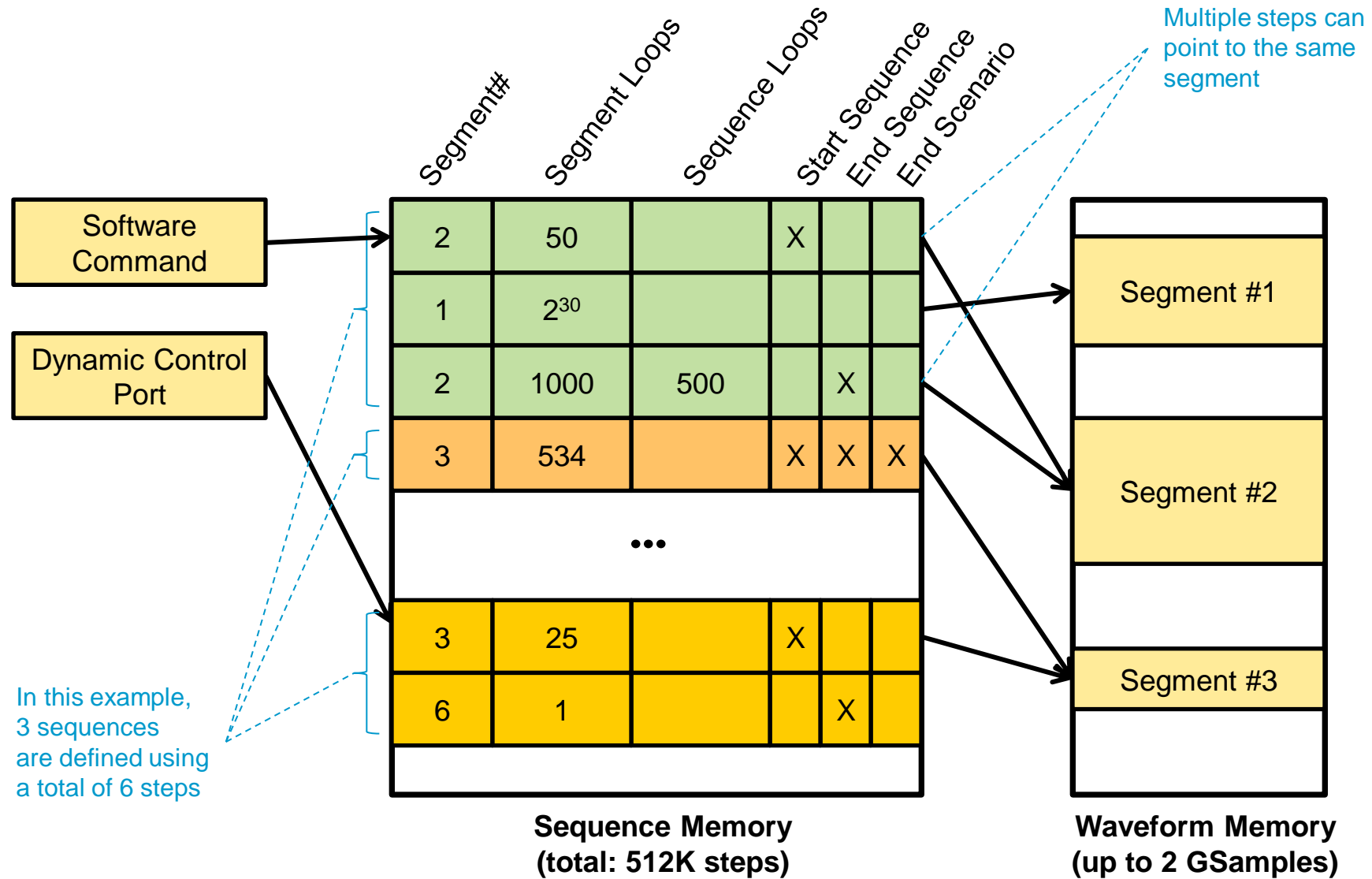
- Start-of-Sequence and End-of-Sequence frame the sequence
- Data entries refer to a waveform segment
- Idle entries implement a sample-accurate delay with a defined sample value
- Advance modes determine how execution proceeds from one entry to the next
- Loop counts determine number of repetitions per segment



Sequence Ids are indices in the sequence table starting at 0.

Segment Ids refer to segments in waveform memory, but not to their memory addresses. They start at 1 and don't need to be ascending.

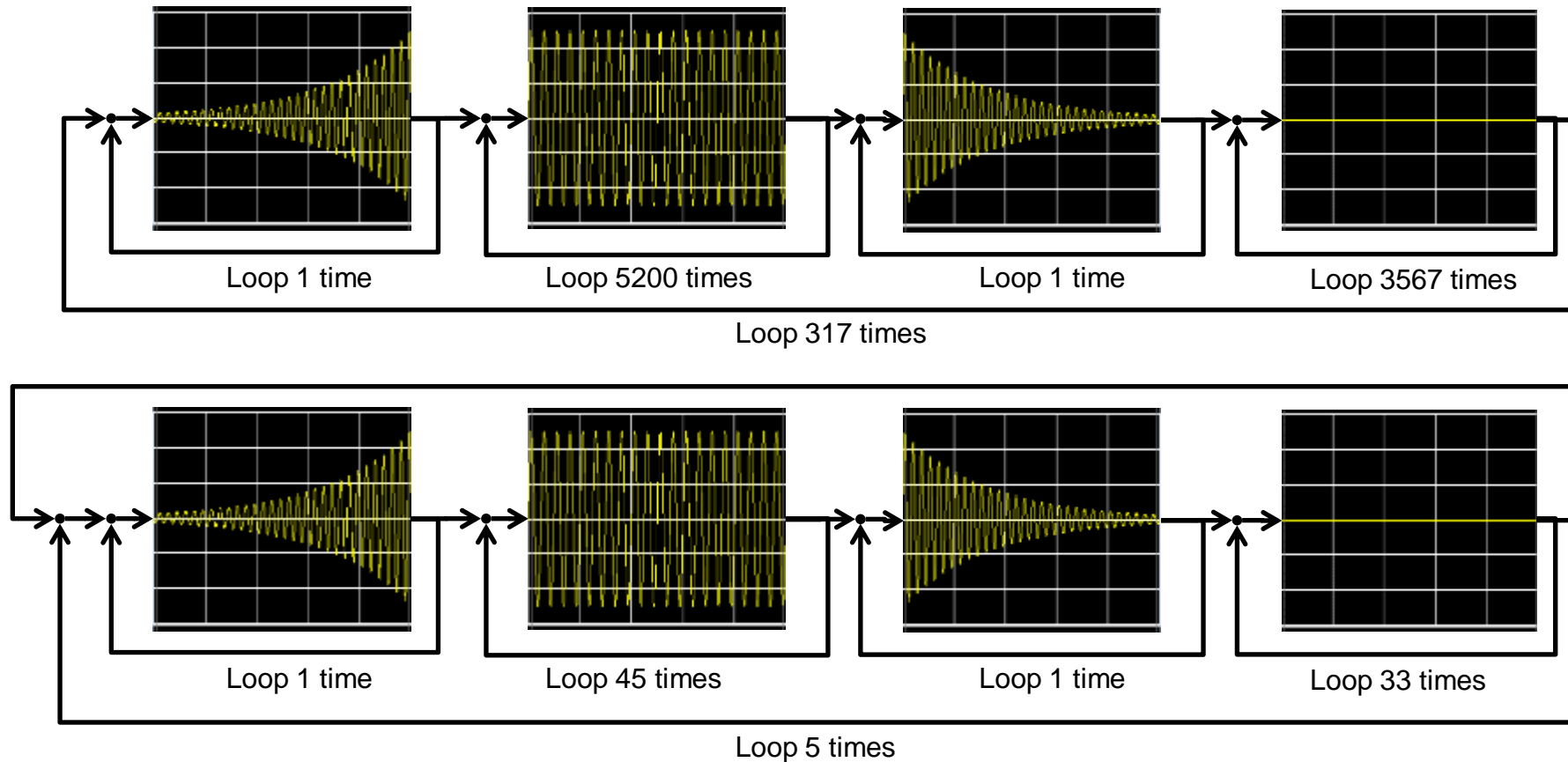
# ...and how does it work internally?



# AWG with Sequencing

Using Scenarios to increase playback time

A **scenario** consists of a list of sequences

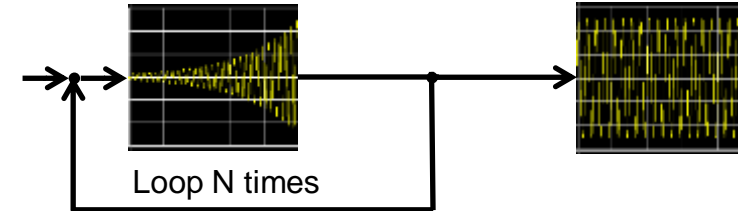


# Advancement Modes of Sequencing

Advancing from one segment/sequence to the next can be...

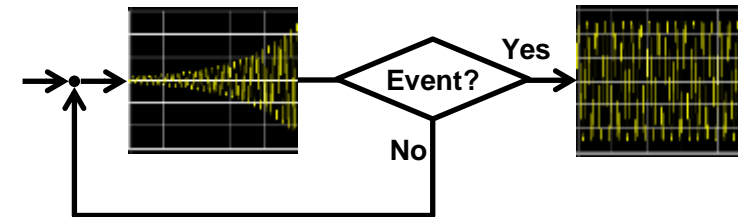
- **Automatic**

- Loop N times, then go to next segment/sequence (un-conditional)



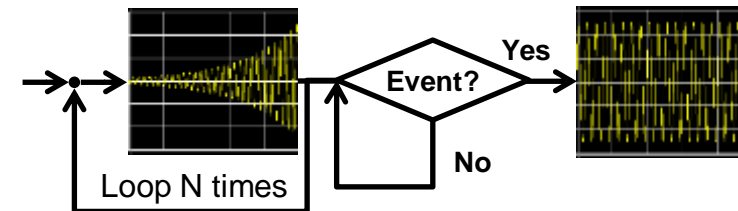
- **Conditional**

- Loop until an event occurs, then go to next segment/sequence



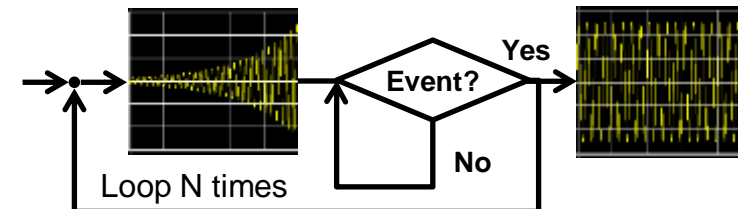
- **Repeat**

- Loop N times, then wait until an event occurs before going to the next segment/sequence

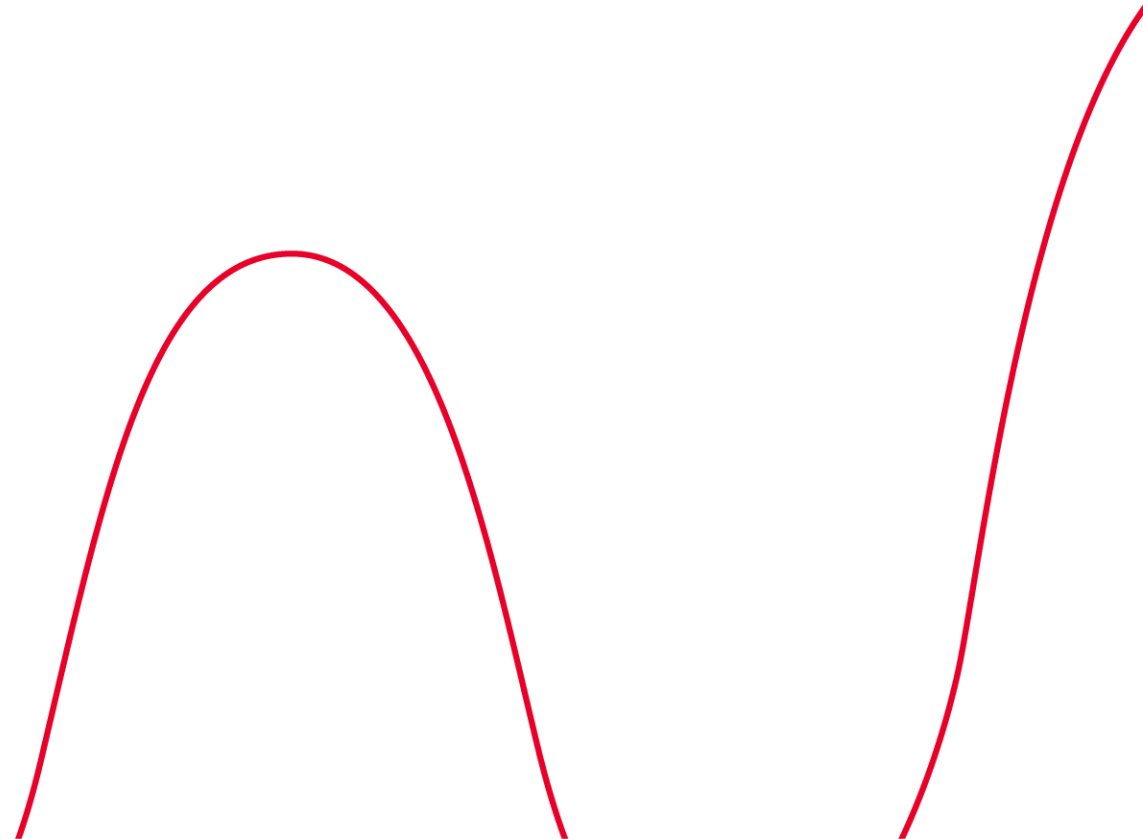


- **Single**

- Same as "Repeat", but wait for an event on every loop



**It's time for a demo...**

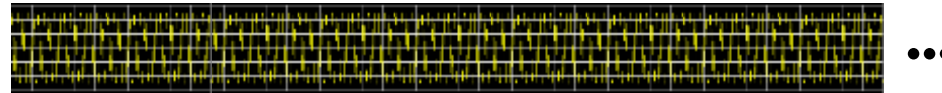




# Trigger Modes of Sequencing

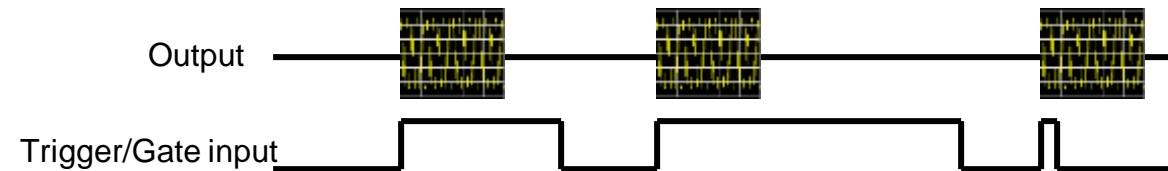
All of the previously mentioned cases can be combined with the following trigger modes. This applies to **segments** or **sequences**.

- Continuous



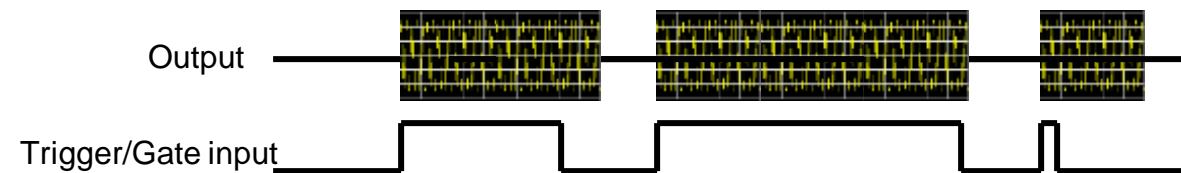
- Triggered

- Each edge of the trigger signal starts the selected segment/sequence



- Gated

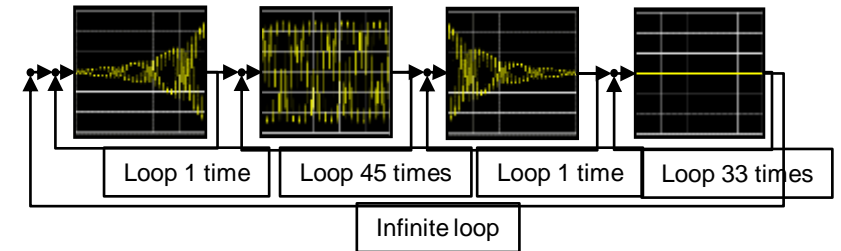
- Segments are always completed



# Selection of segment/sequence to be generated

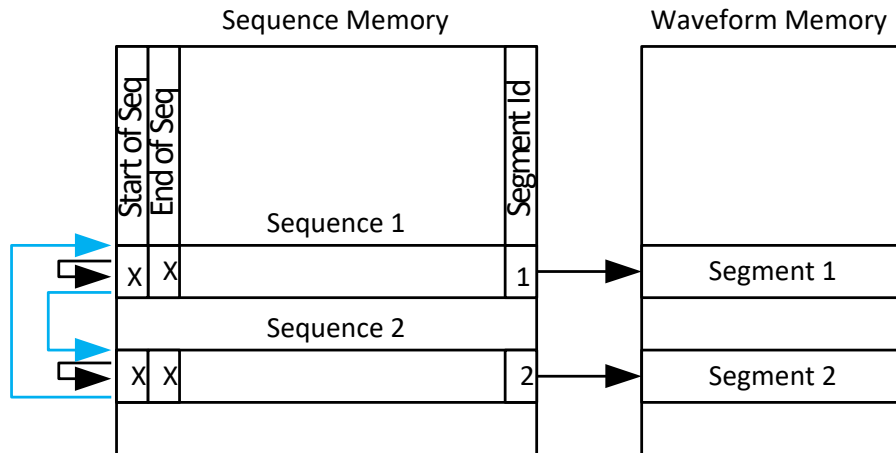
Selection of segment/sequence can be determined by...

- **Pre-defined sequence**
  - If the **order** of waveform segments is known ahead of time, it can be set up as a “sequence”
- **Dynamic Control Port**
  - The dynamic control port on the front panel allows you to select one of  $2^{13}$  ( $2^{19}$ ) segments/sequences **dynamically at runtime** by applying a digital pattern to the dynamic control port connector
- **Software**
  - Instead of applying a digital pattern to the dynamic control port, you can also select a segment/sequence using software by sending a command to the firmware



**In all cases, transitions are “seamless” - without any gaps**

# Sequencing – “Memory Ping-Pong”



Execute alternately Segment 1 and 2. When one segment is played the other one is loaded with a new waveform.

Load Segment 1. Start signal generation in dynamic mode. → Segment 1 is played (Looped until next is selected or played once in triggered mode).

Repeat:

- Load Segment 2.

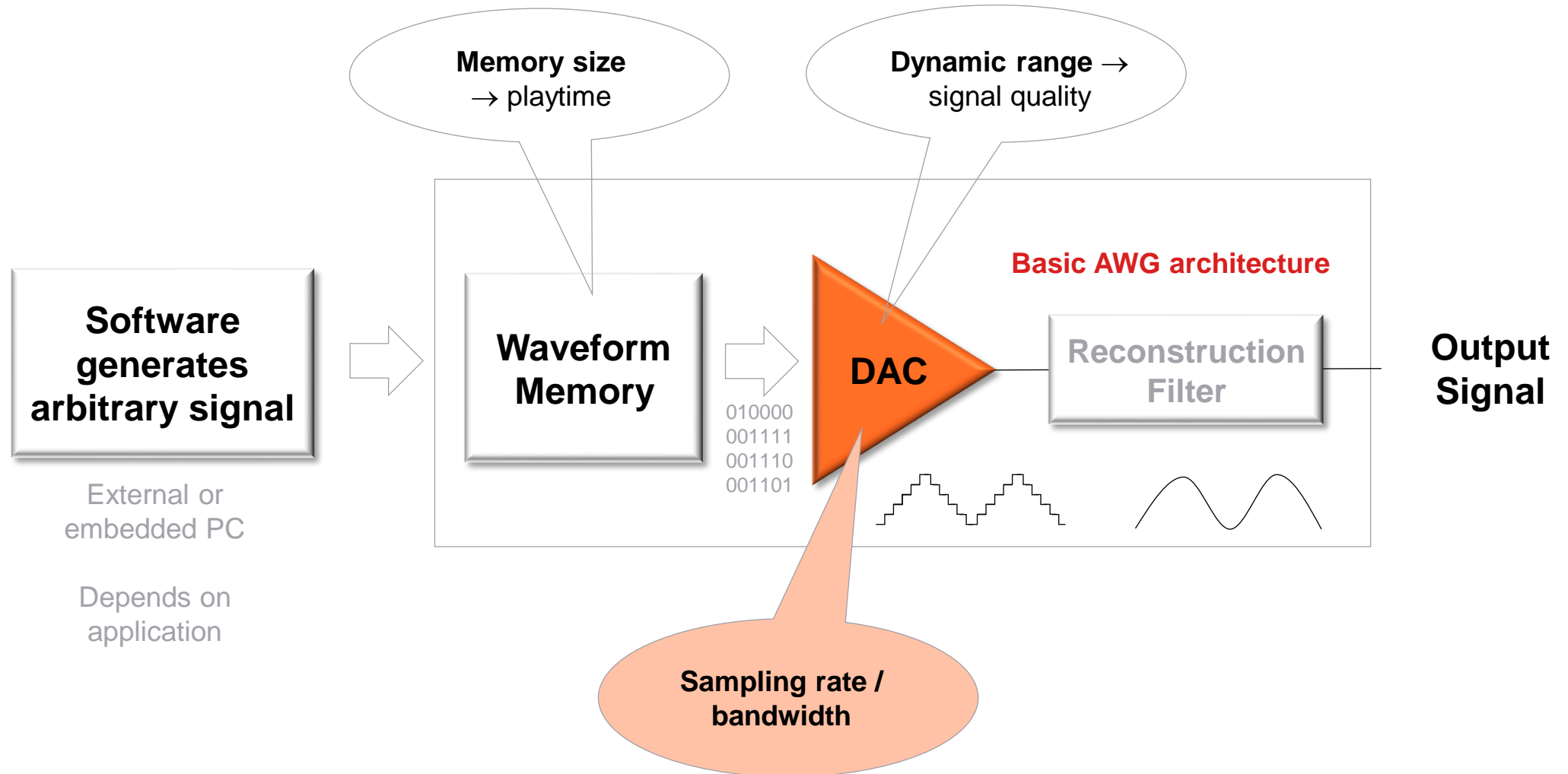
- Switch to play Segment 2.

- Load Segment 1.

- Switch to play Segment 1.

# Theory of AWG Operation

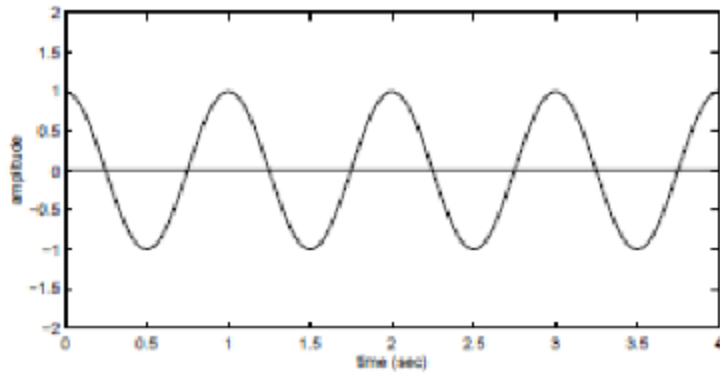
## Key Blocks and Specifications: Sampling rate



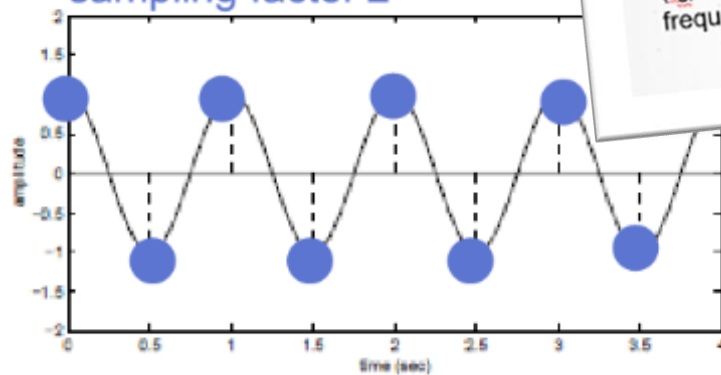
# AWG Sampling Basics

## Sampling Factor

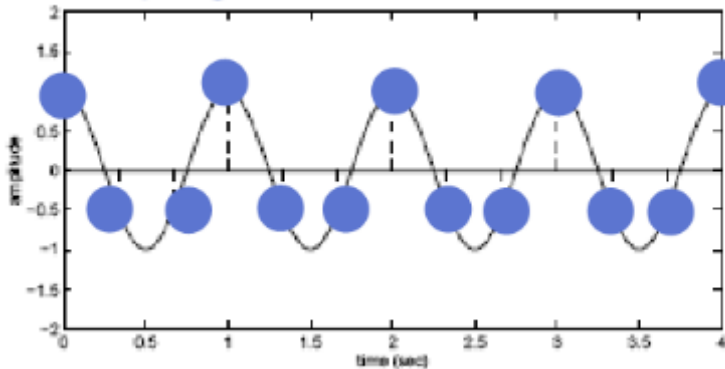
Sample Rate determines **Modulation Bandwidth** = typically less than 1/2 of sample rate, e.g. 12 GSa/s  $\rightarrow$  5 GHz BW



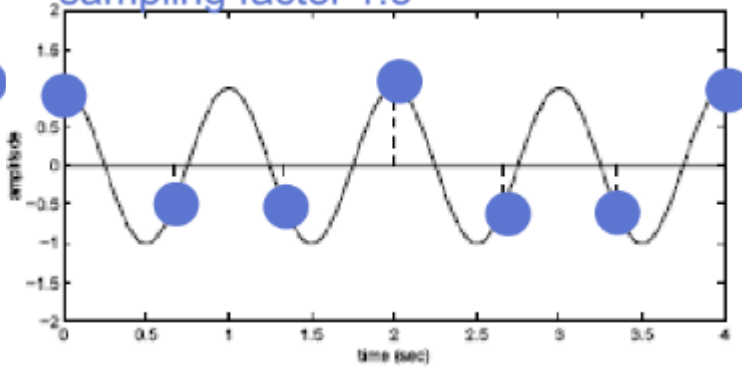
sampling factor 2



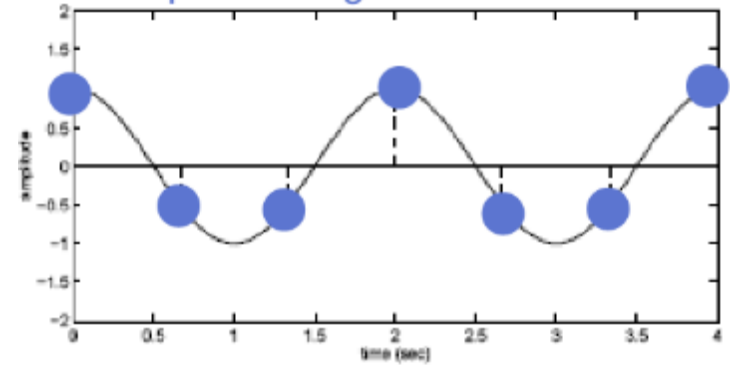
sampling factor 3



sampling factor 1.5



example aliasing



## Sampling Basics

NYQUIST'S THEOREM: REMEMBERING THE UNIVERSITY DAYS

- Nyquist's sampling theorem states that for a limited bandwidth (band-limited) signal with maximum frequency  $f_{max}$ , the equally spaced sampling frequency  $f_s$  must be greater than twice of the maximum frequency  $f_{max}$ , i.e.,

$$f_s > 2 \cdot f_{max}$$

in order to have the signal be uniquely reconstructed without aliasing.

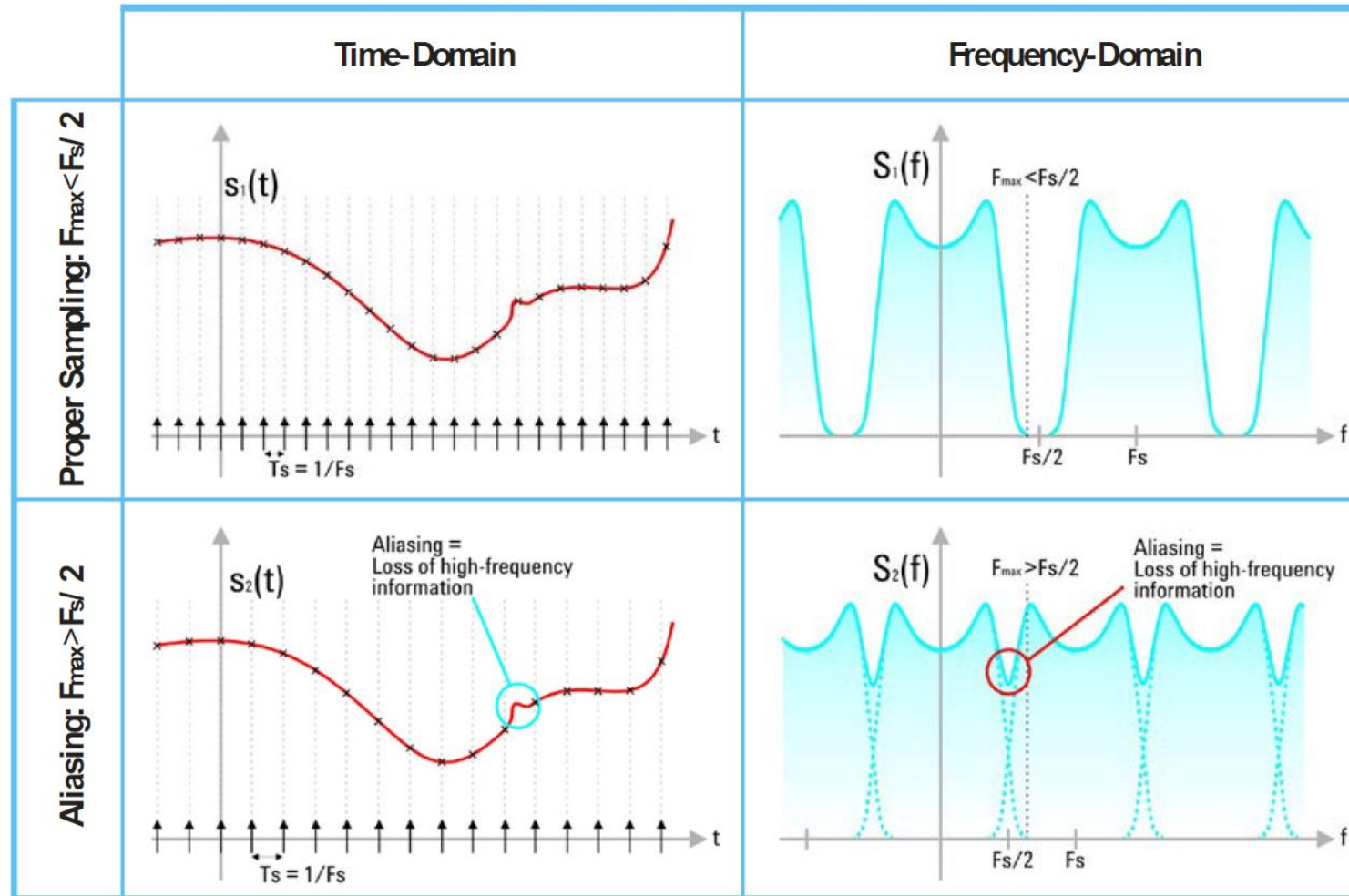
- The frequency  $2 \cdot f_{max}$  is called the Nyquist sampling frequency ( $f_N$ ). Half of this value,  $f_{max}$ , is sometimes called the Nyquist frequency ( $f_N$ ).



Dr. Harry Nyquist, 1889-1976, articulated his sampling theorem in 1928

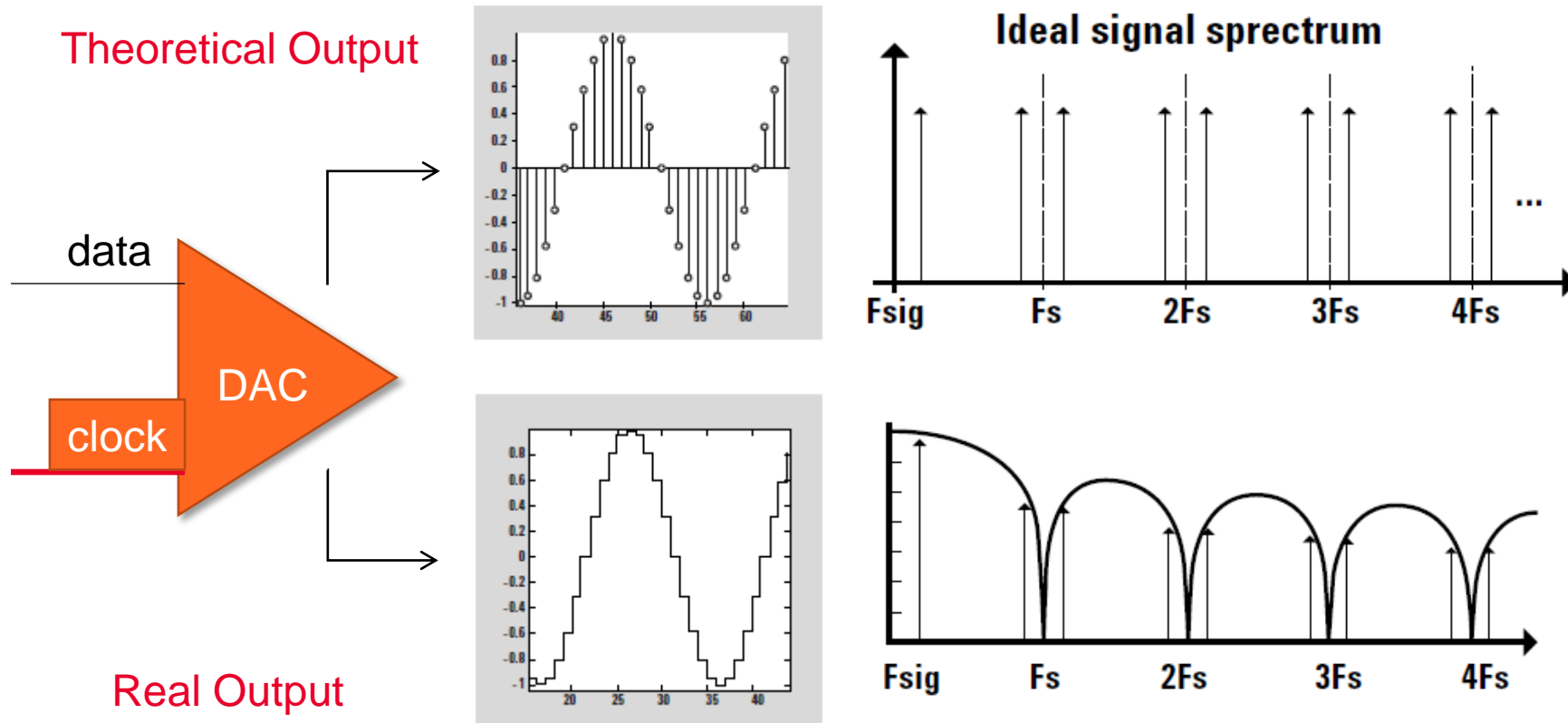
# Nyquist Theorem in the Time and Frequency Domains

## Nyquist Sampling Theorem



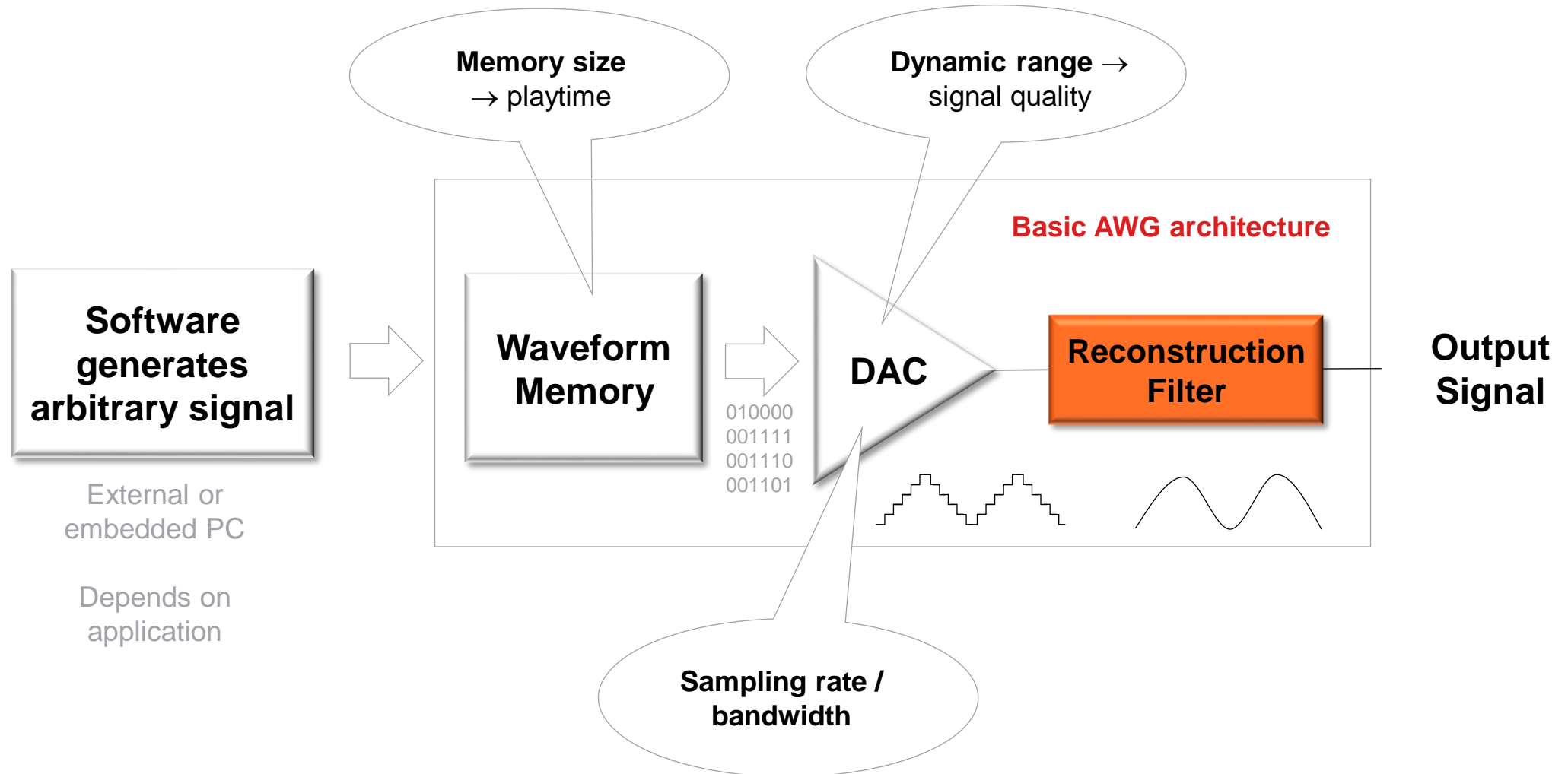
# AWG Sampling Basics

Theoretical vs. Real output



# Theory of AWG Operation

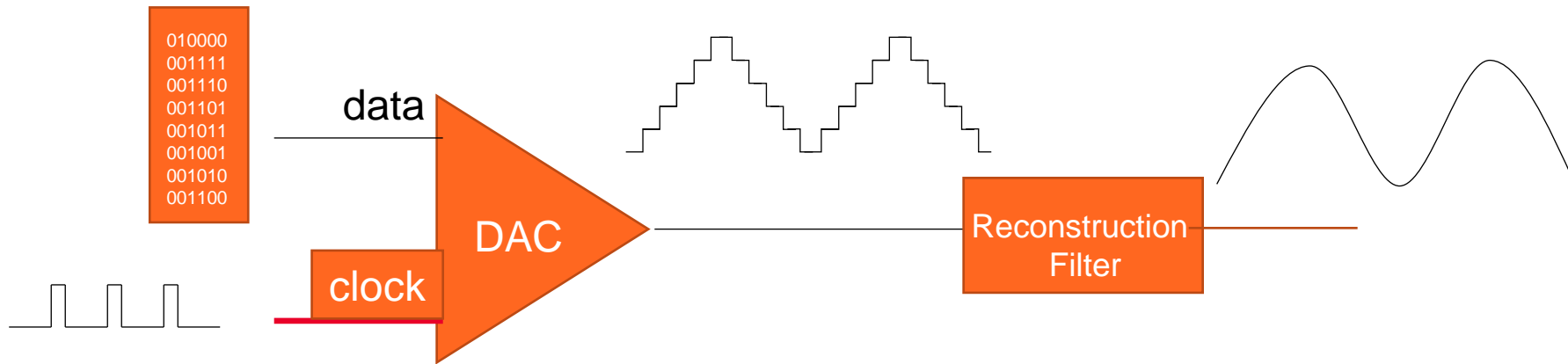
## Key Blocks and Specifications: Reconstruction Filter



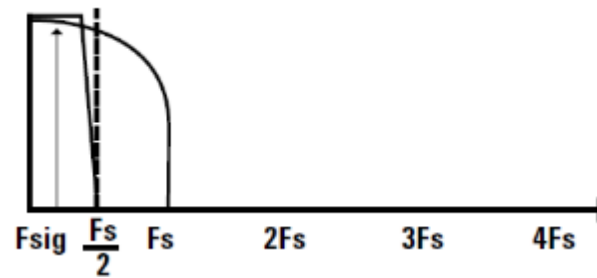
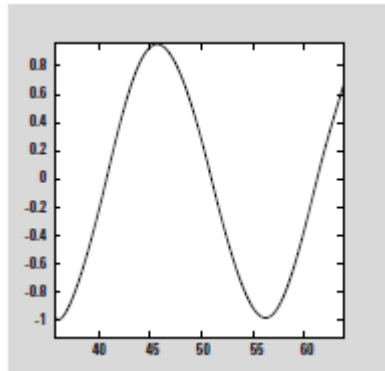


# AWG Sampling Basics: Reconstruction Filter

## Theory of AWG operation

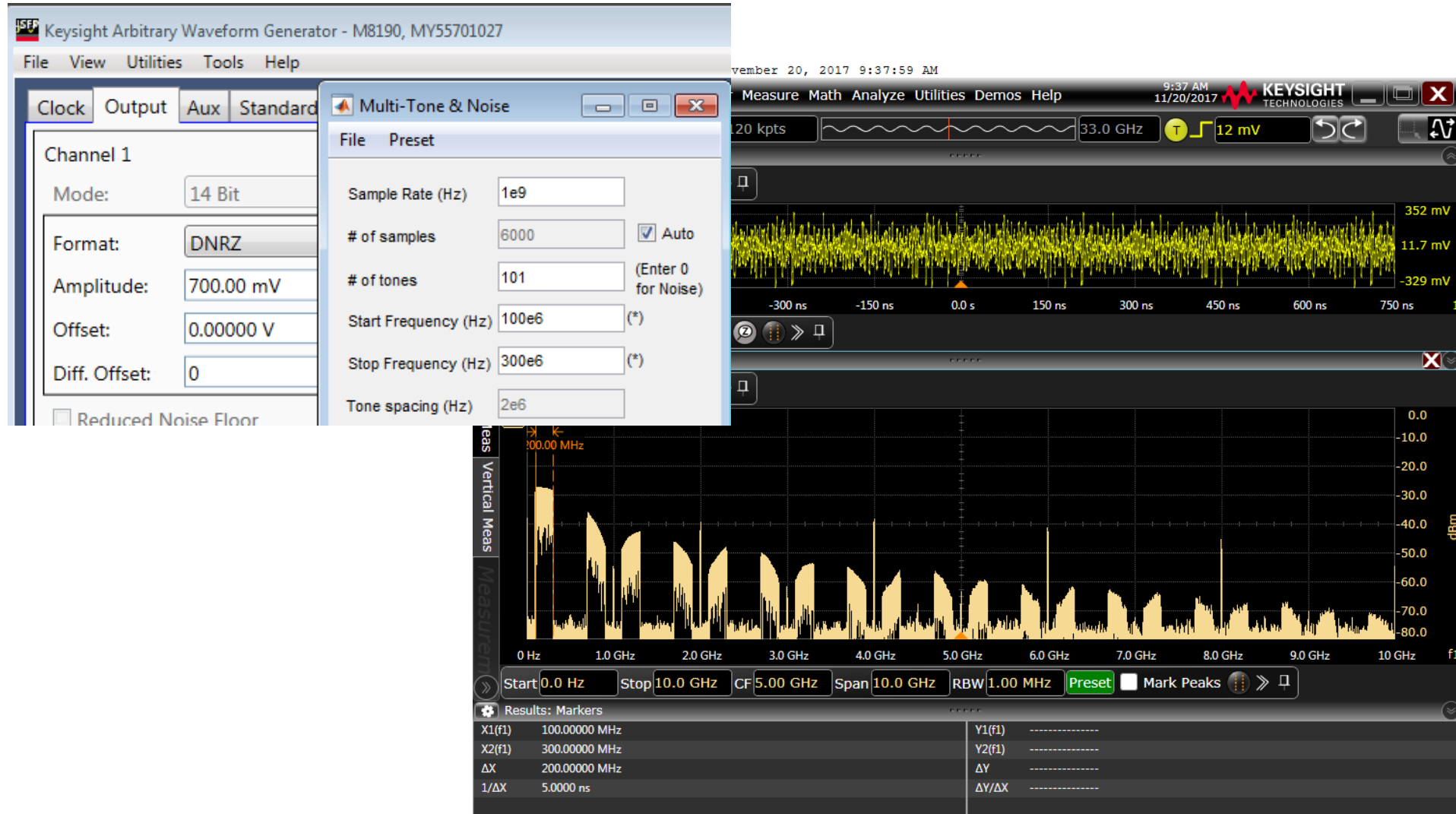


010000  
001111  
001110  
001101  
001011  
001001  
001010  
001100

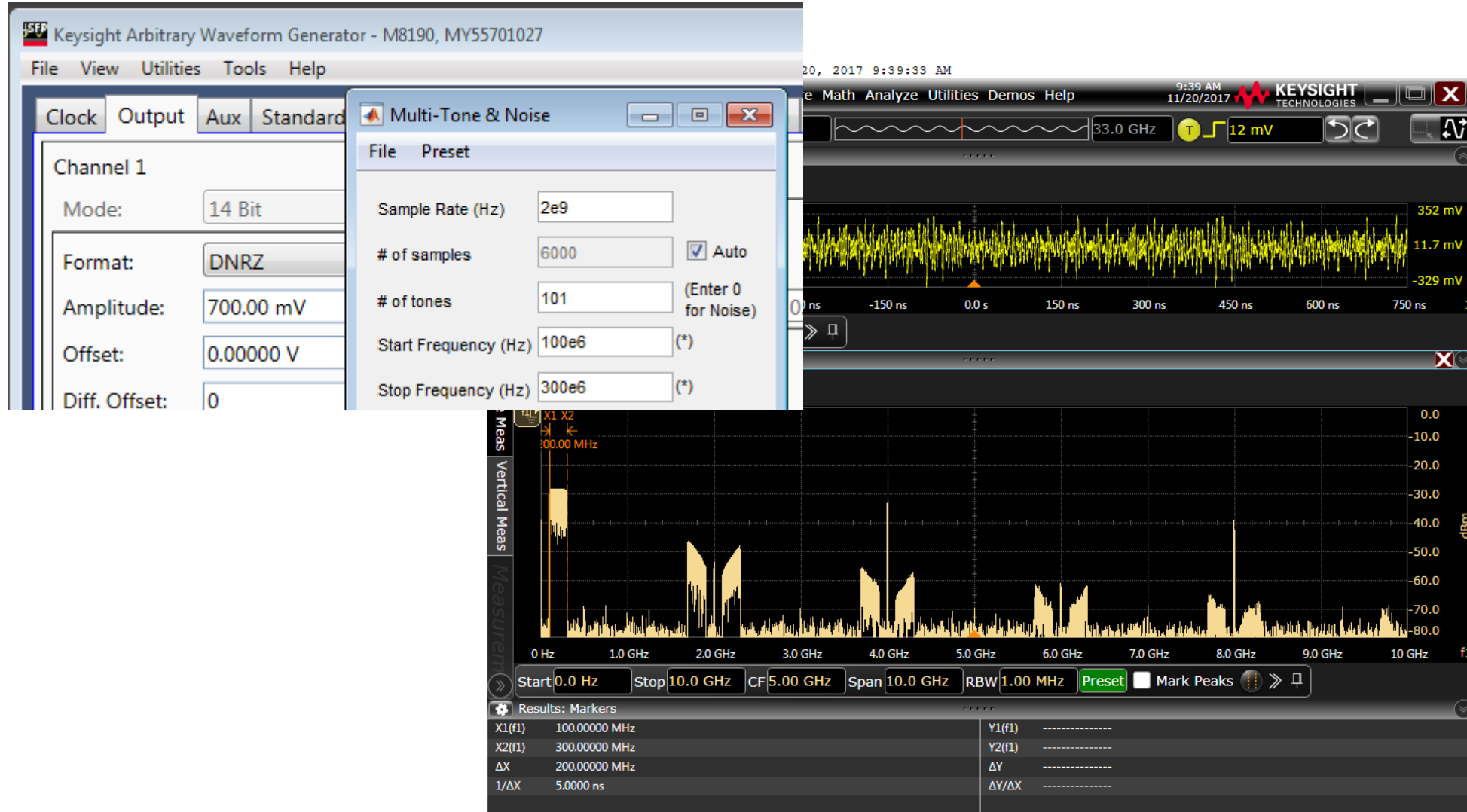


**Why is a high sampling frequency  
always good?**

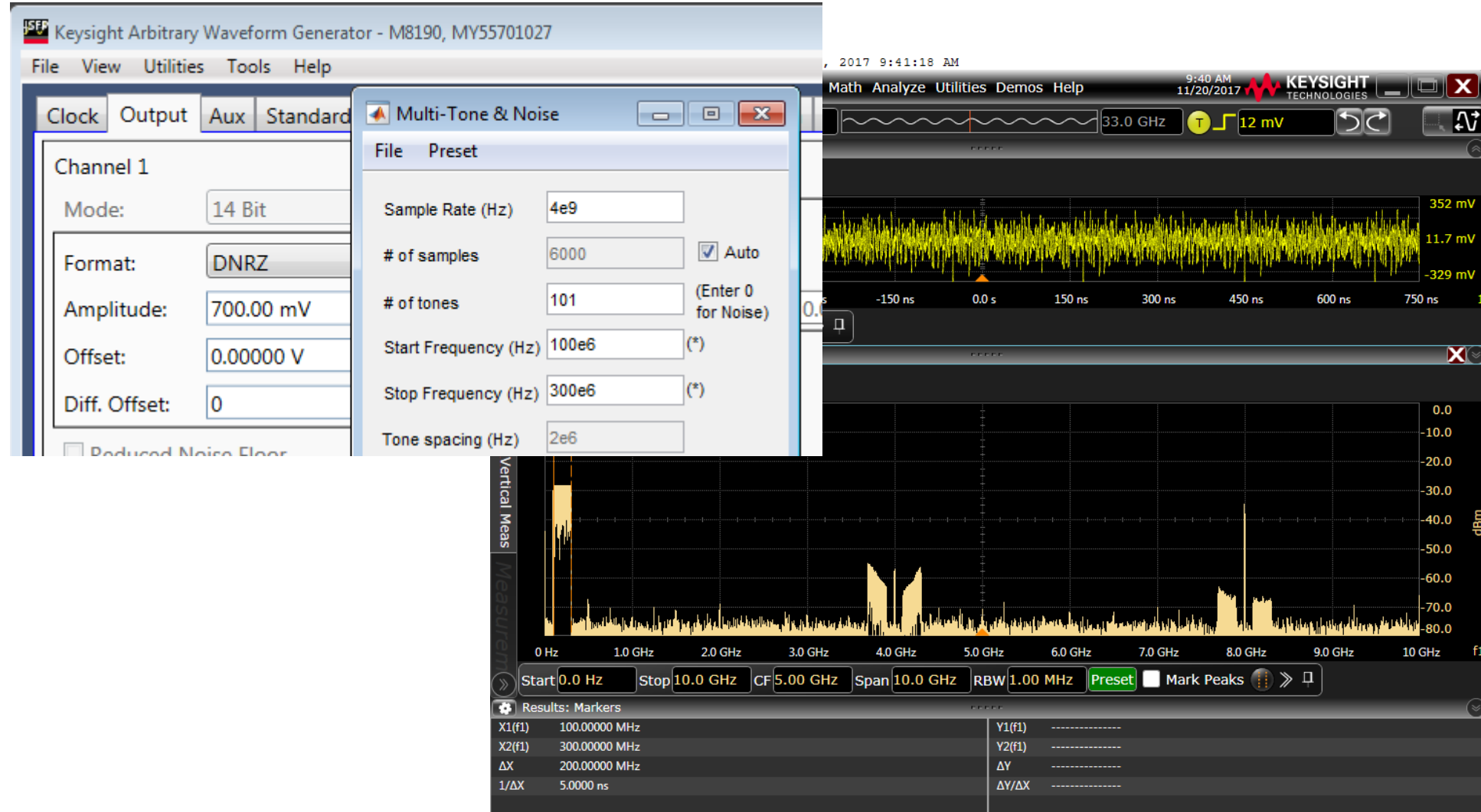
# 100 to 300 MHz multitone & 1 GSample/s sample rate



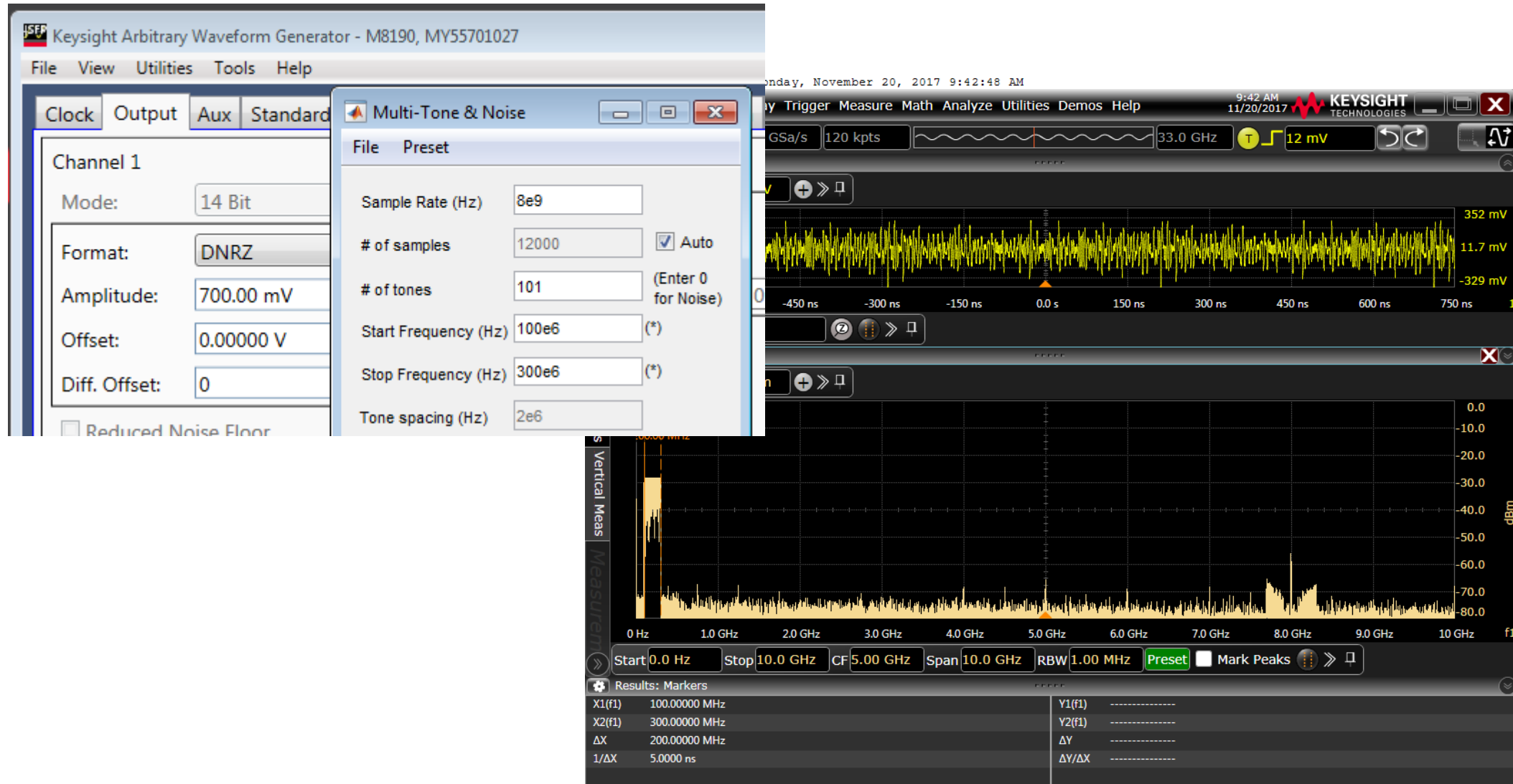
# 100 to 300 MHz multitone & 2 GSample/s sample rate



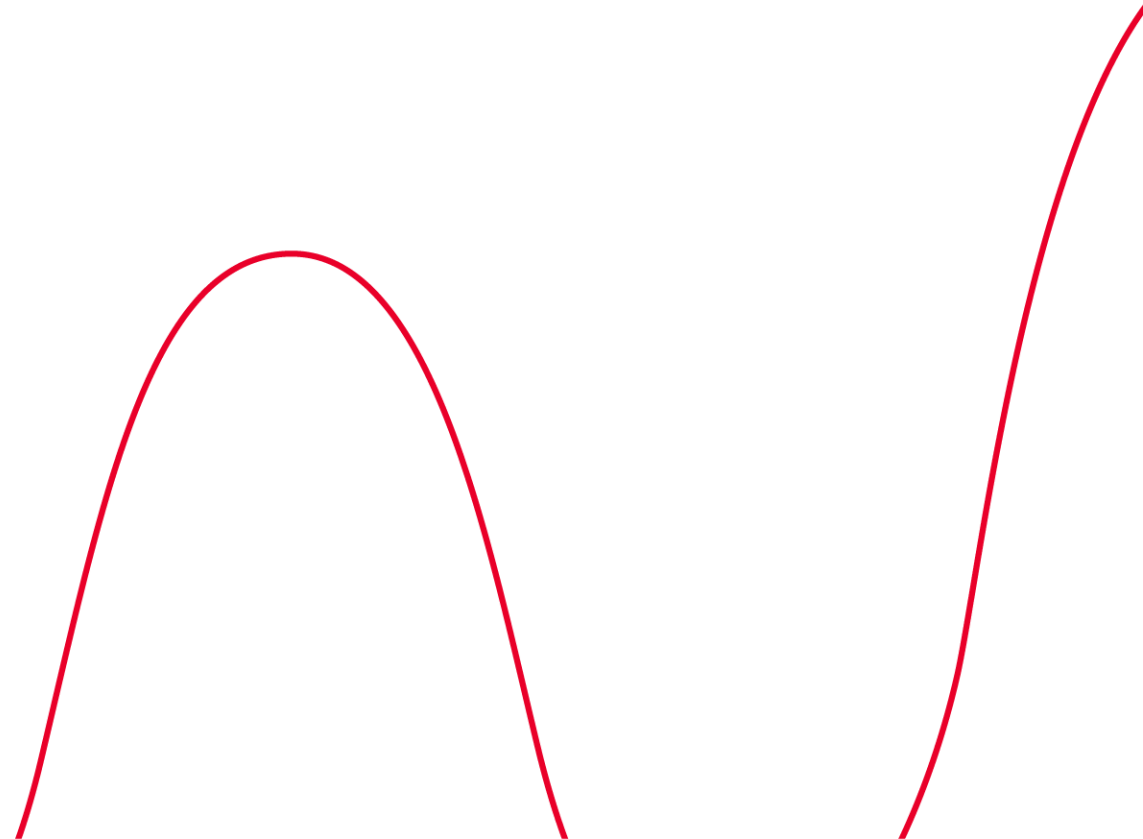
# 100 to 300 MHz multitone & 4 GSample/sec sample rate



# 100 to 300 MHz multitone & 8 GSa/s sample rate

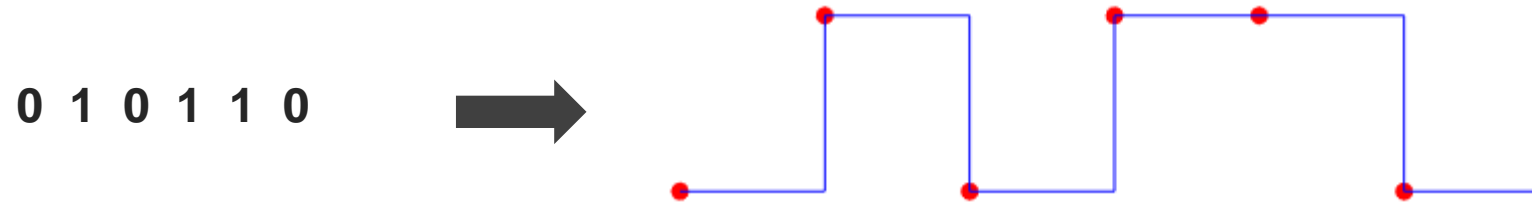


**It's time for a demo...**

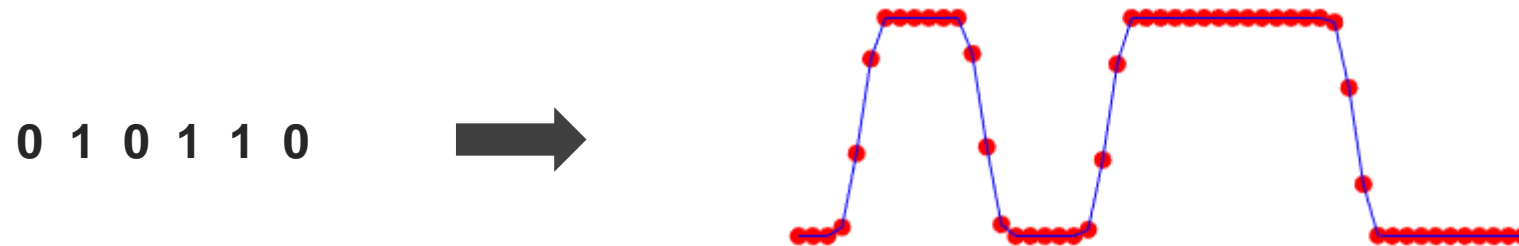


# How To Get From a Bit/Symbol Pattern To Samples

- In a “one-sample-per-symbol” architecture (e.g. in a BERT-PG), it is straight forward:

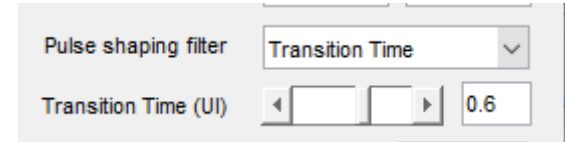


- In an AWG, you typically have more than one sample per symbol and the ratio (sample rate / symbol rate) is no necessarily an integer



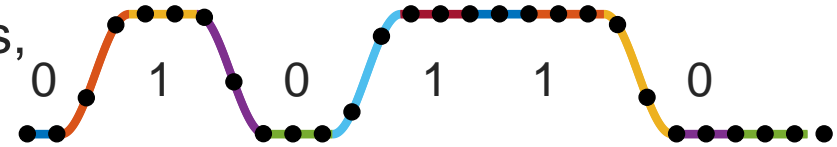


# Algorithm #1 – Variable Transition Time



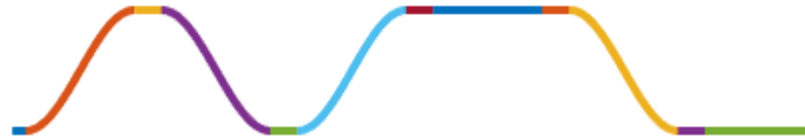
How to get from symbols to samples

- Combine “straight line” and “cosine” waveform segments, then place sampling points on the calculated shapes



- Depending on the desired rise and fall time, the cosine segments are „squeezed“ or „expanded“

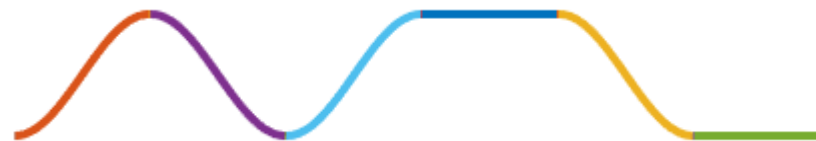
- Slow transition time:



- Fast transition time:



- With transition time = 1 UI, the pulse shape is very close to a Raised Cosine with alpha = 1 (i.e. sinusoidal)



- **This algorithm works well with large oversampling ratios (i.e. low baudrates)**

# Algorithm #2 – Using a Pulse Shape Filter

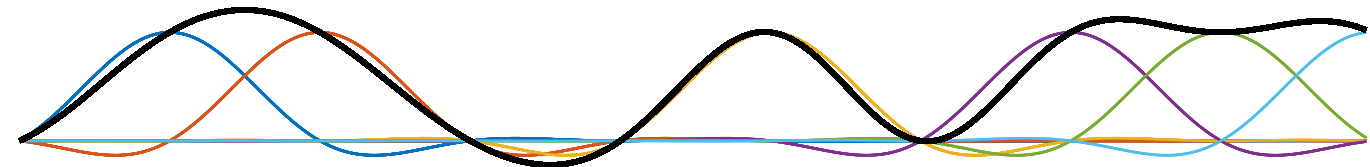


How to get from symbols to samples

- For RF signals, a **pulse-shape filter** is usually applied in order to reduce the occupied bandwidth. The same approach can be used for NRZ or PAM-n signals to reduce their bandwidth
- Each symbol is treated as a dirac pulse, convoluted with impulse response of the pulse shape filter, typically a raised cosine or root-raised-cosine
- Example: impulse response of a raised cosine filter with  $\alpha = 0.3$
- The final waveform is **the sum of the impulse responses of all symbols**



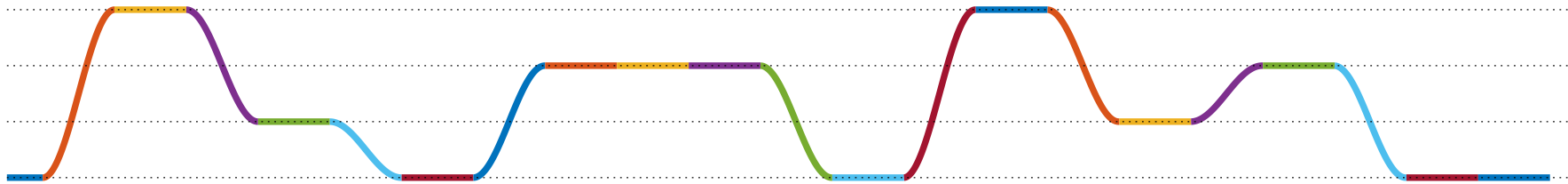
1 1 0 0 1 0 1 1 1



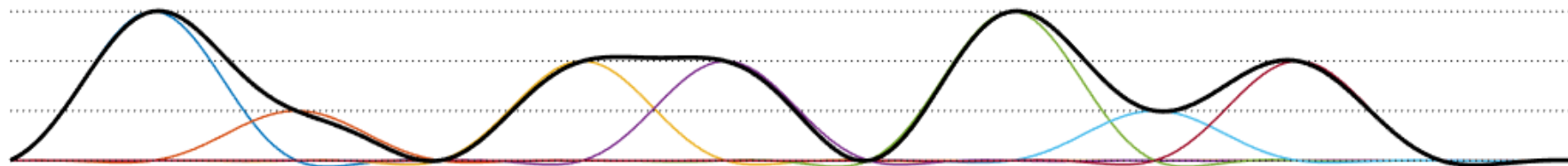
# Multi-Level signals (PAM-n)

## How to get from symbols to samples

- Both previously mentioned algorithms work with multi-level (PAM-n) signals equally well. Here is an example with PAM-4:
- Transition time algorithm



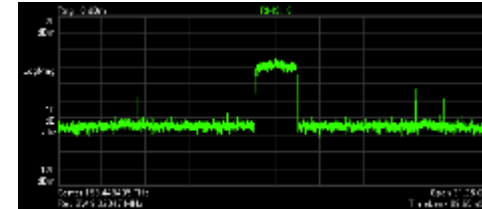
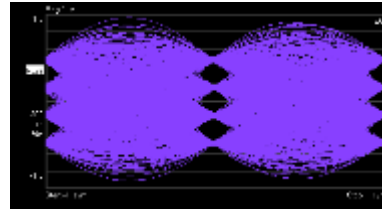
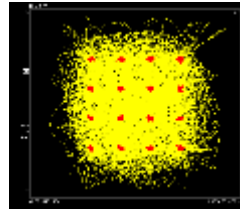
- Pulse-Shape algorithm



# Spectral Effects of Pulse Shaping

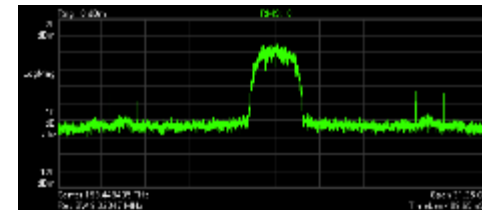
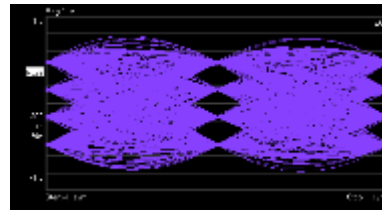
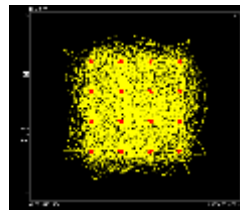
Pulse Shape affects Signal Waveform

Narrowest spectrum



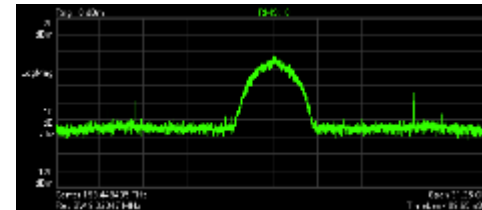
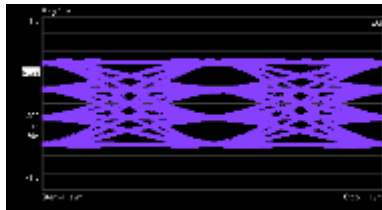
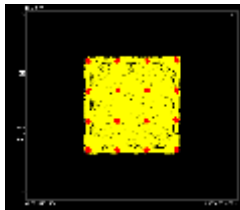
Raised cosine  
 $\alpha = 0.05$

Best EVM



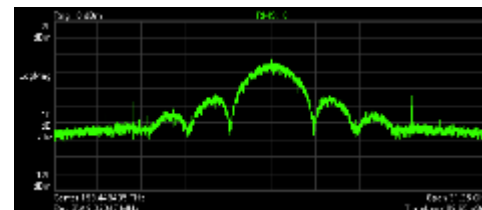
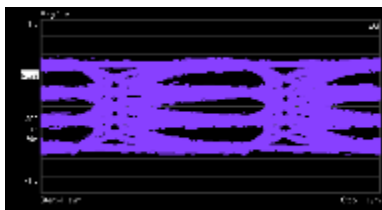
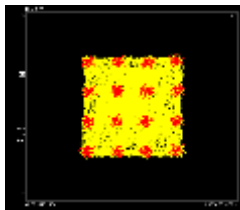
Raised cosine  
 $\alpha = 0.35$

Best Q-Factor



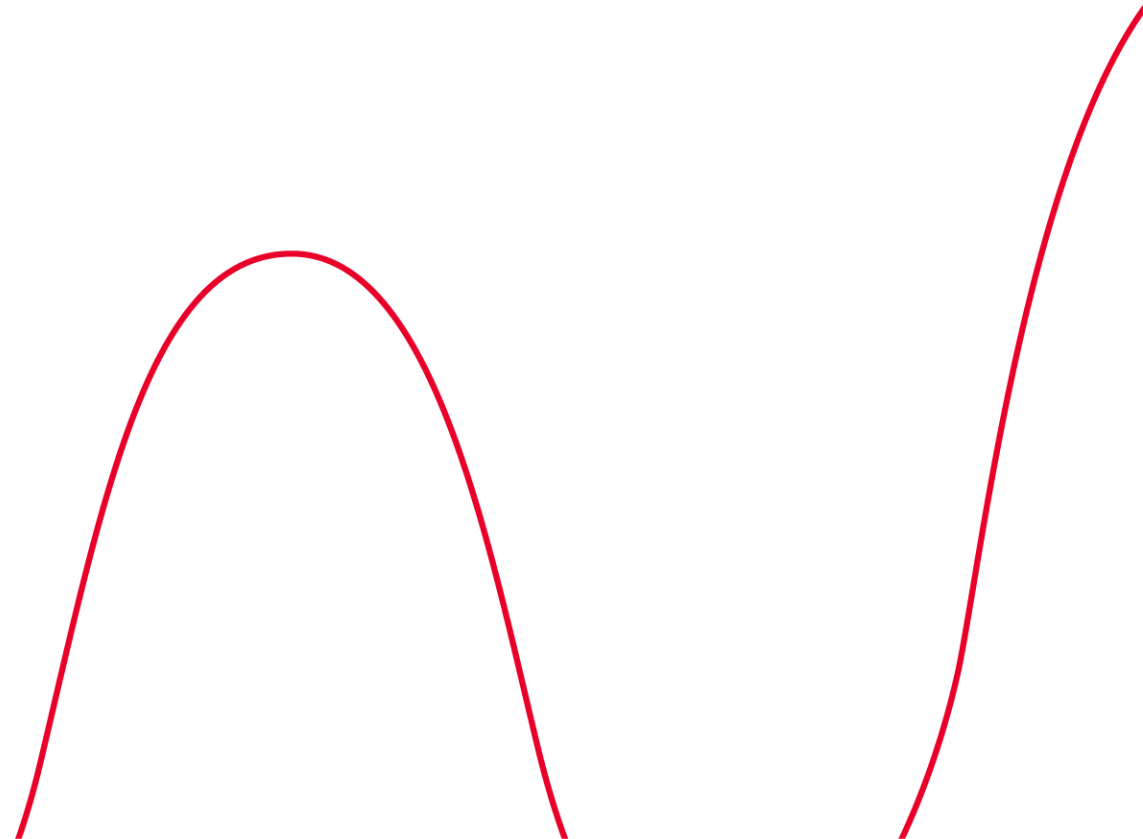
Raised cosine  
 $\alpha = 1.0$

Lowest Jitter



Transition time = 0

**It's time for a demo...**



# Pattern Definitions

- **Random**

- uses the MATLAB “rand()” function to generate random bits/symbols
- Works also for “odd” PAMs (e.g. PAM-3, PAM-5, PAM-6, etc.) where there is no  $2^n$  mapping from bits
- is not limited to a certain number of symbols and hence works well for demo purposes ;-)

- **PRBS  $2^n-1$**

- Standard PRBS patterns, can be used with NRZ or PAM4
- Be aware that the correct pattern length will likely require the sample rate to be adjusted

- **SSPRQ, QPRBS-13, etc.**

- Special patterns for PAM-4 testing. Same restrictions as with PRBS patterns

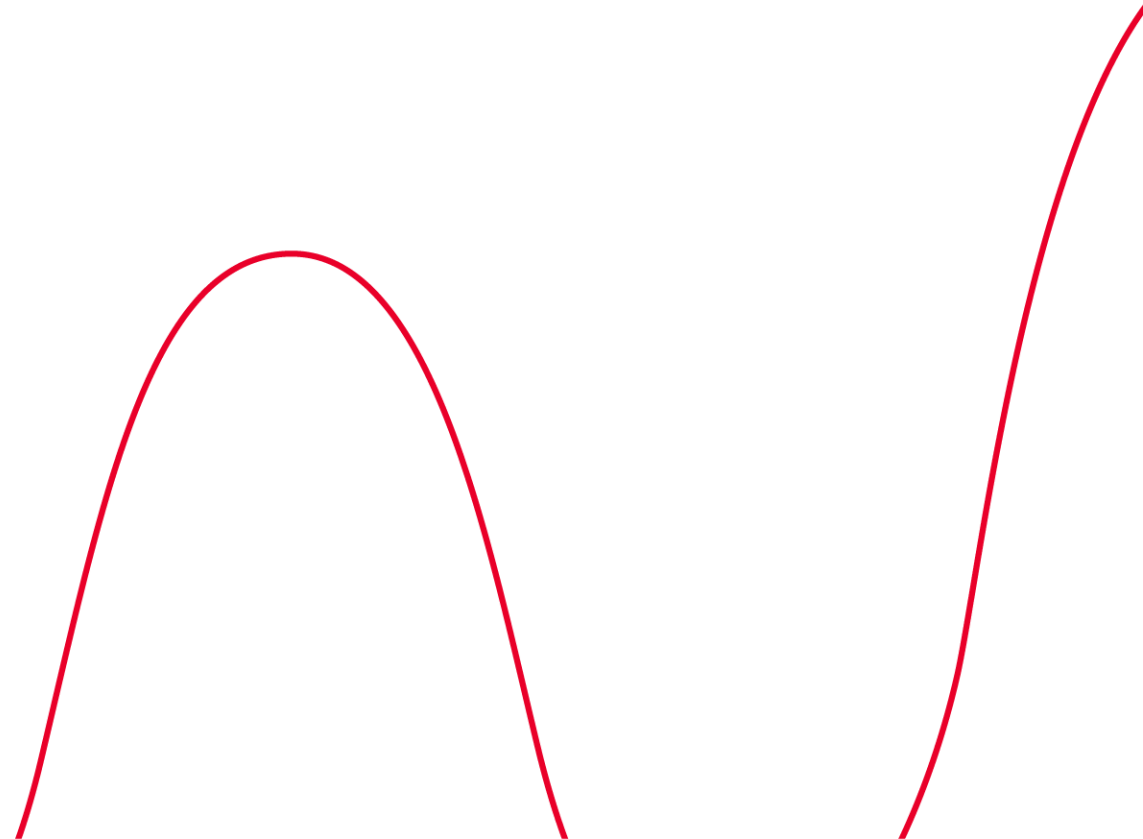
- **User defined**

- Allows an arbitrary MATLAB expression to be specified as a pattern
- E.g. `[randi([0 3],1,128]` will generate 128 random PAM-4 symbols. You can use variable names and functions from the MATLAB workspace, e.g. `csvread()`

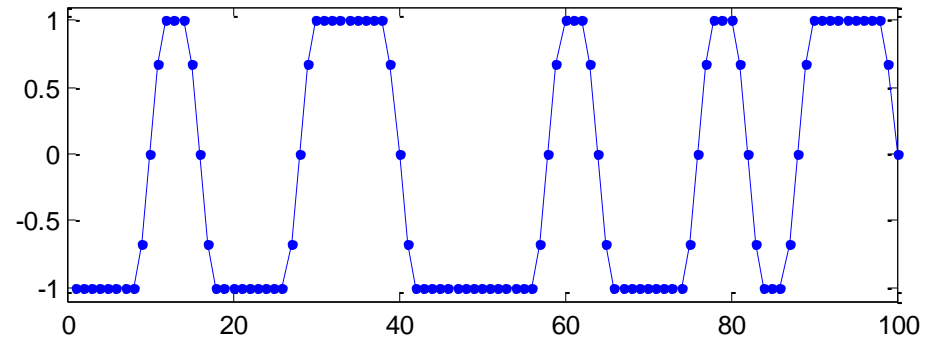
- **Pattern from file**

- Reads symbol values from a file. Values can be either 0 and 1 for NRZ or 0,1,2,3 for PAM4 or even decimal values for slight offsets from nominal signal levels (e.g. 0 1 0 1 0 **0.9** 0 1 0 1)

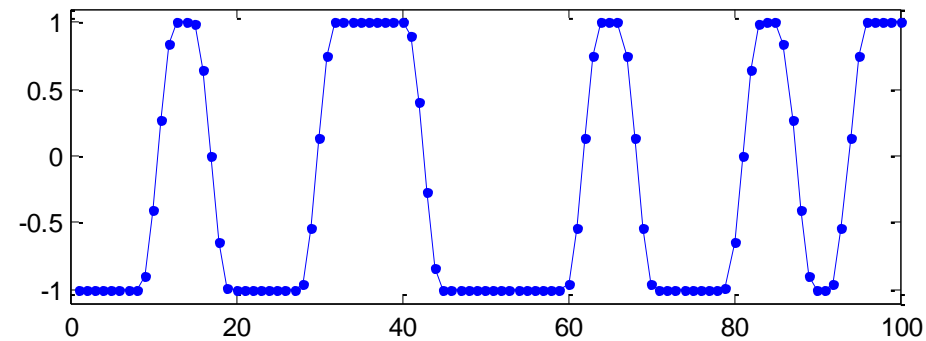
**It's time for a demo...**



# Integer vs. Fractional Re-sampling



Integer Re-sampling



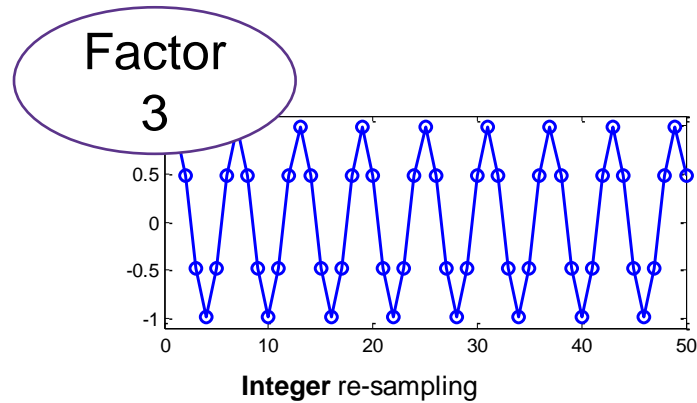
Fractional Re-sampling



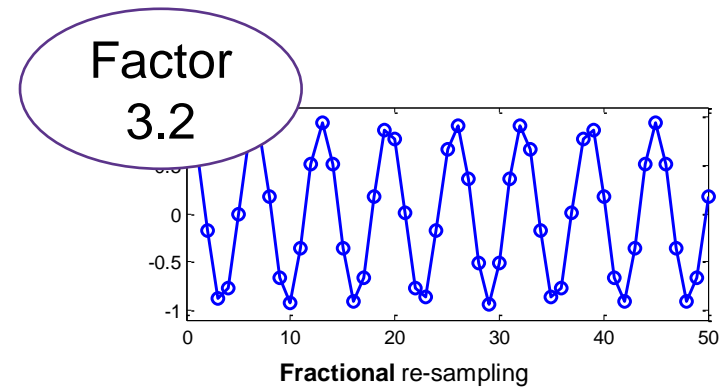
# Integer vs. Fractional Re-sampling

## 32 Gbaud Signals and Beyond

Example: Symbol rate: 20 GSa/s, QPSK, raised cosine,  $\alpha=1$



slightly lower jitter for clean signals, but cannot add infinitesimally small amount of timing distortions

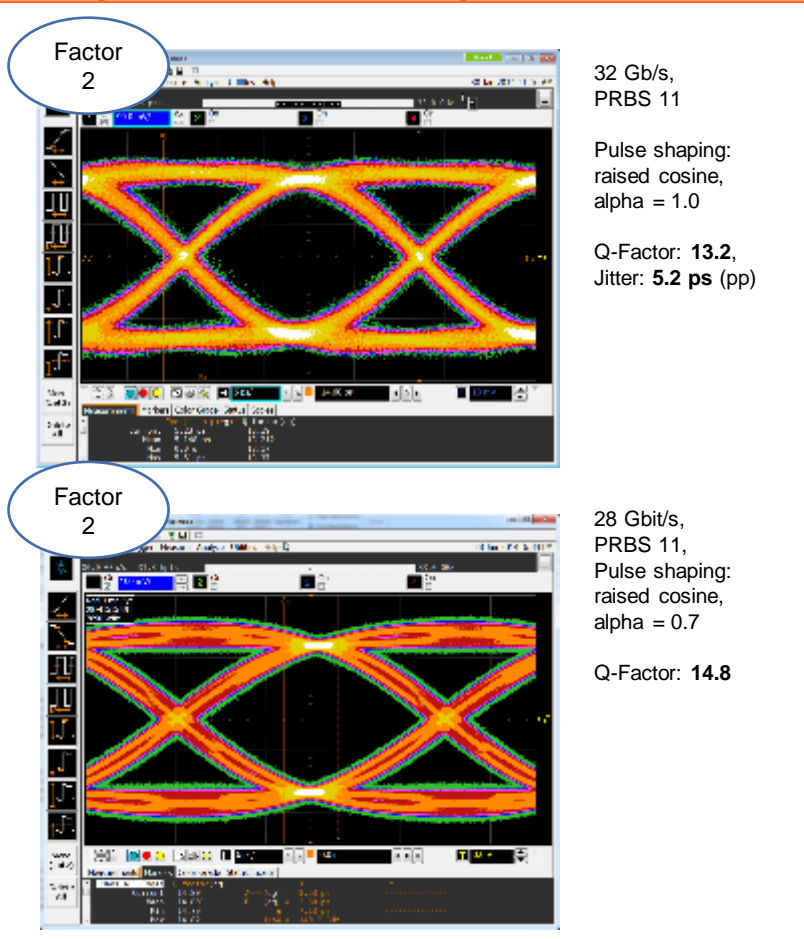


intrinsic jitter slightly higher, but distortions can be added smoothly

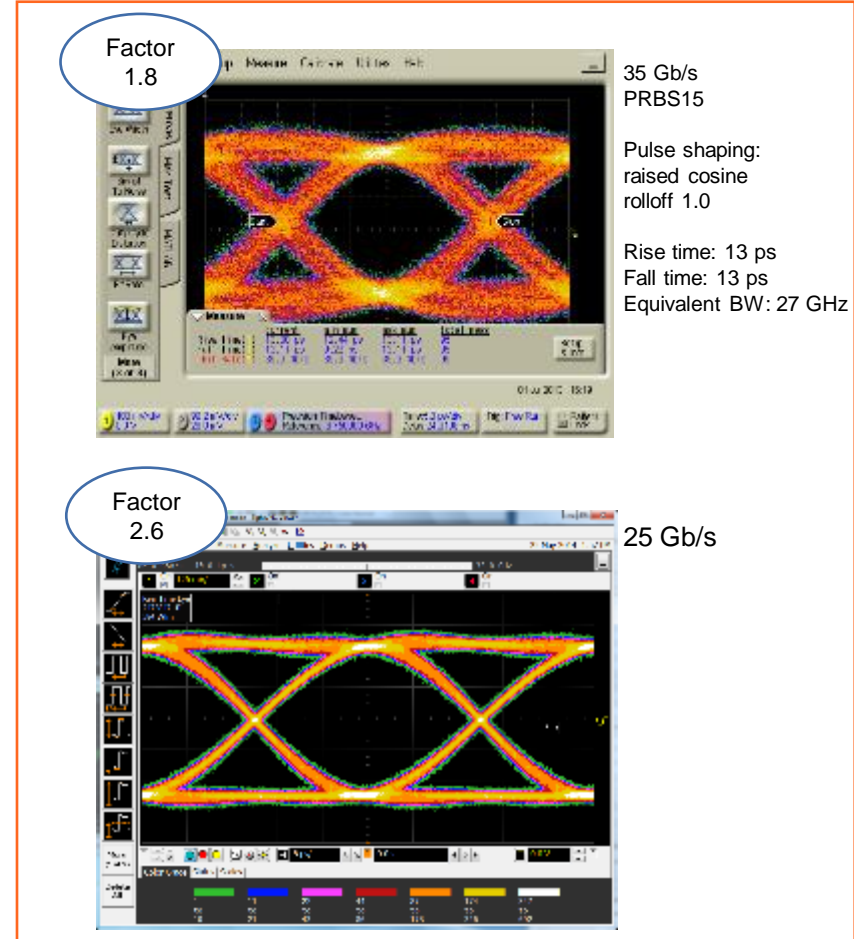
# Integer vs. Fractional Re-sampling

## M8195A Different Sample Factors

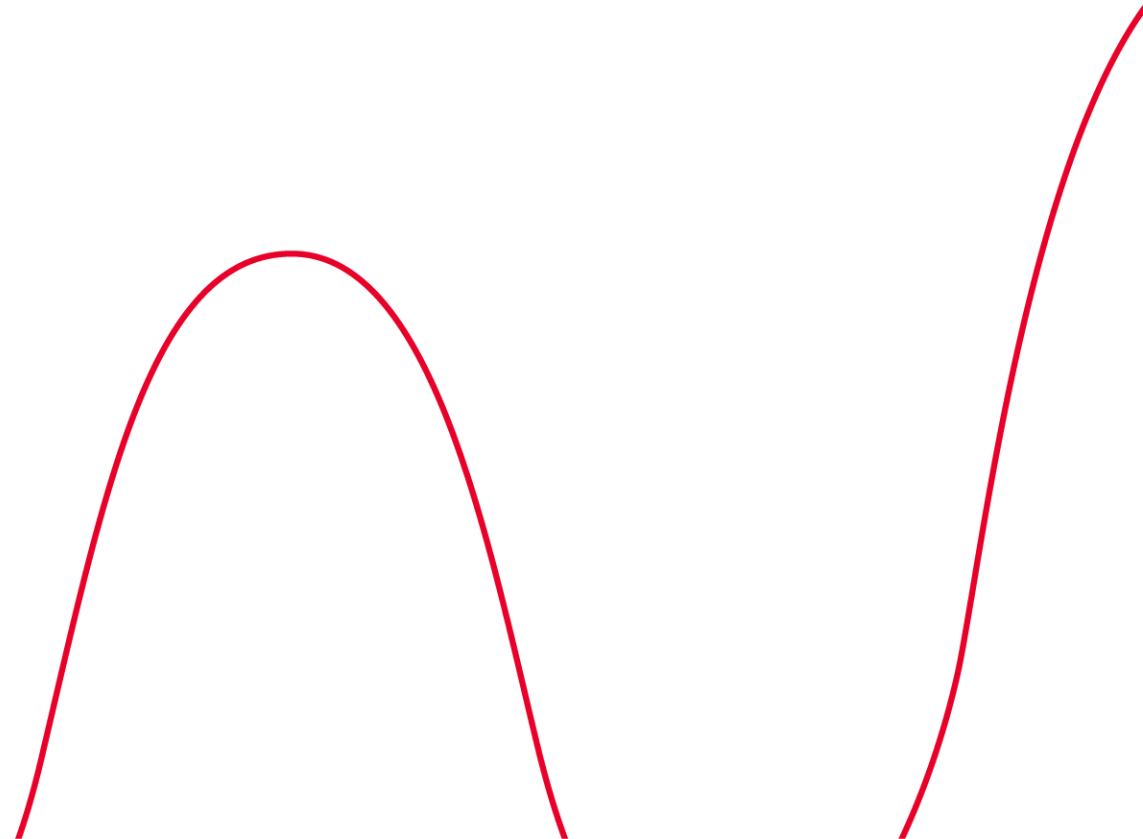
### Integer Re-sampling



### Fractional Re-sampling

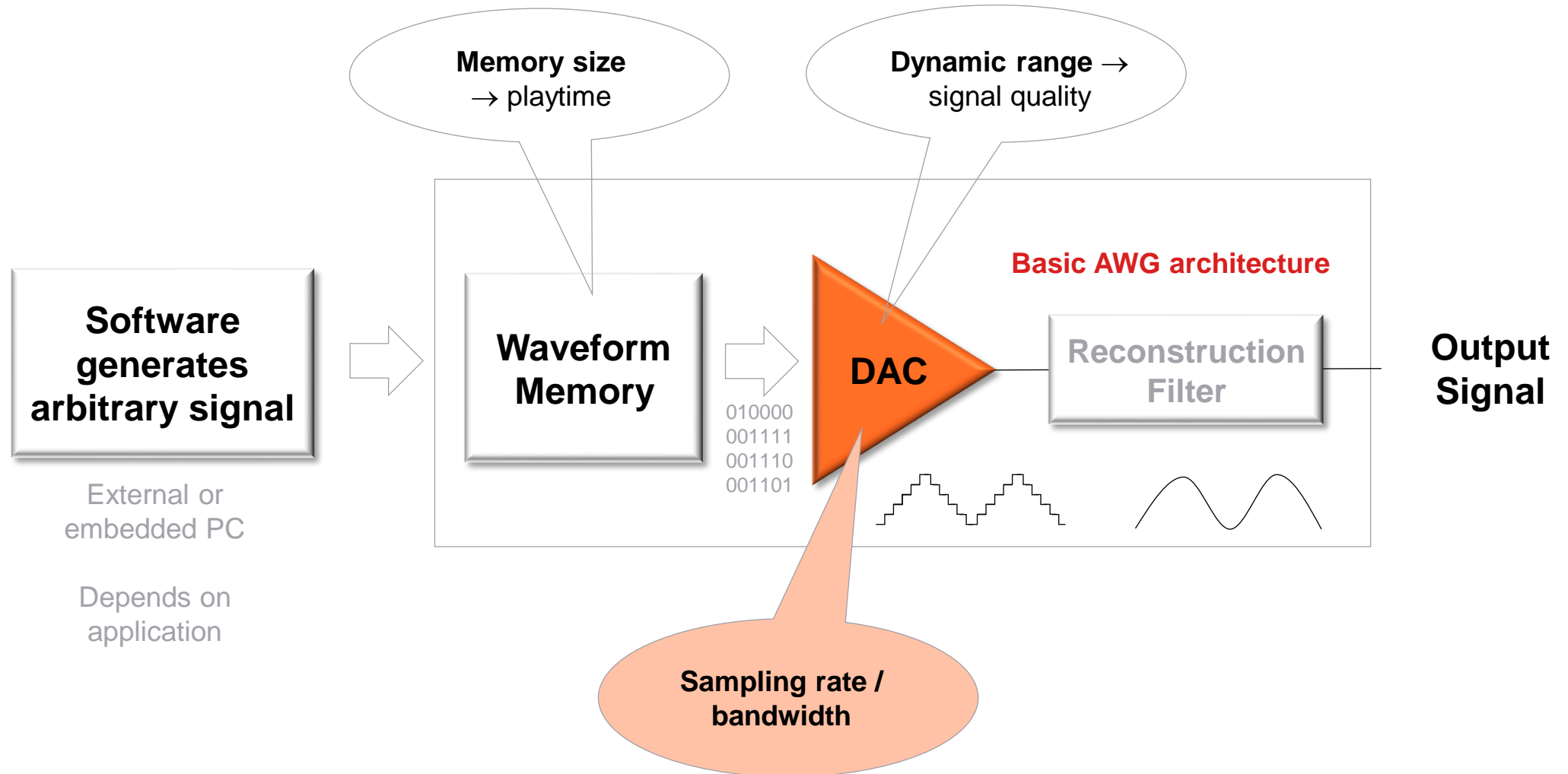


**It's time for a demo...**



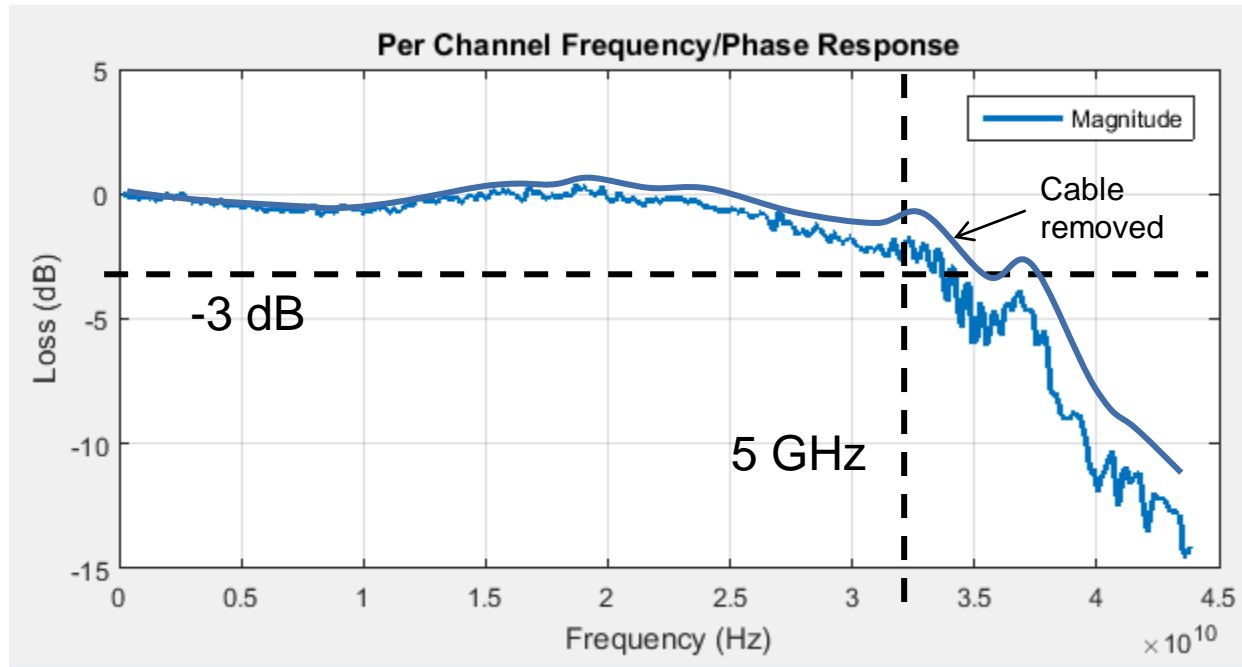
# Theory of AWG Operation

## Key Blocks and Specifications: Bandwidth



# AWG Basics: Analog Bandwidth

## Frequency Response

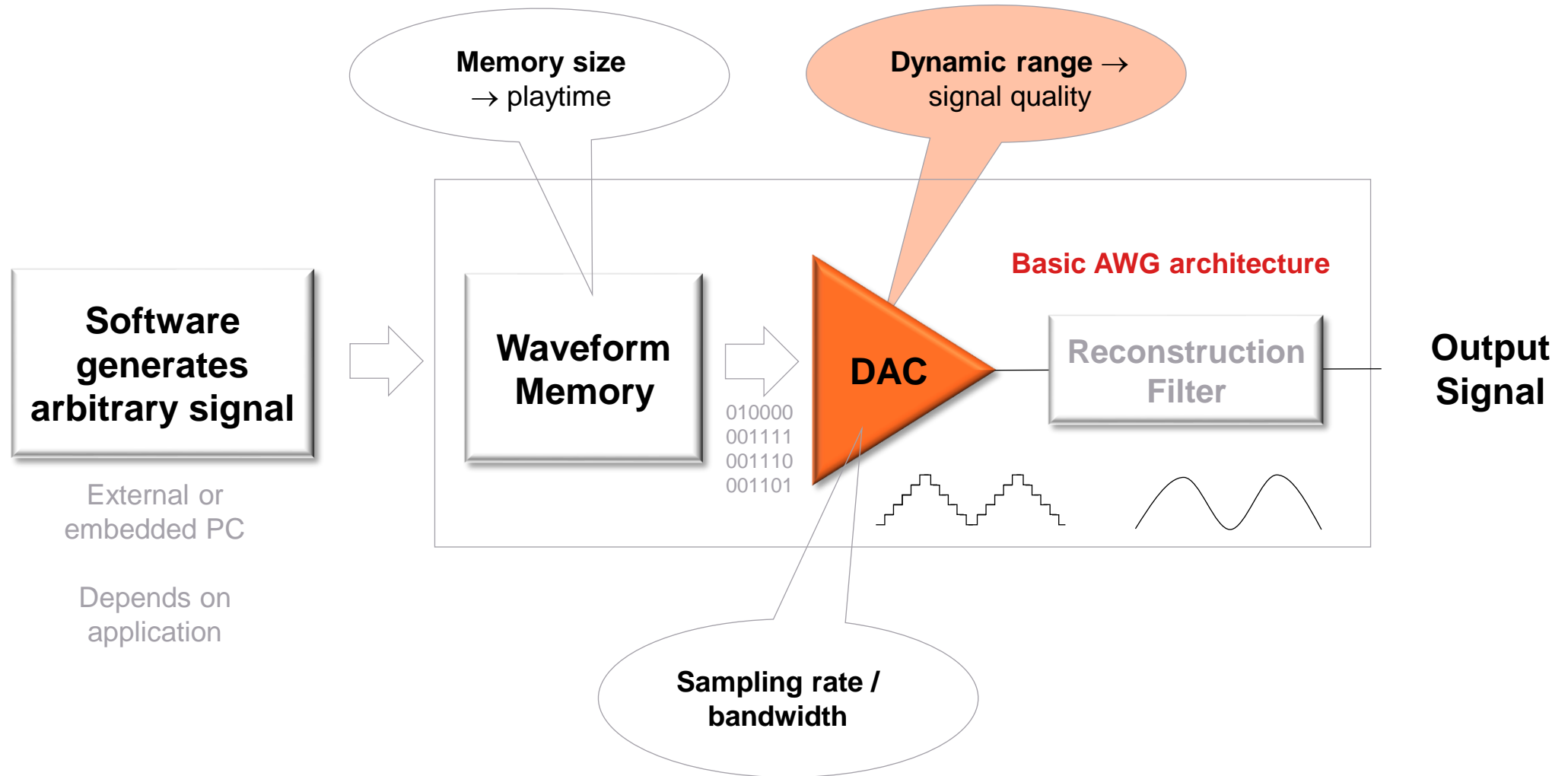


3 dB point

The 3 dB point defines the analog bandwidth

# Theory of AWG Operation

Key Blocks and Specifications: Dynamic Range/signal quality



# AWG Basics: Dynamic Range and Resolution

## Quantisation and Quantisation Error

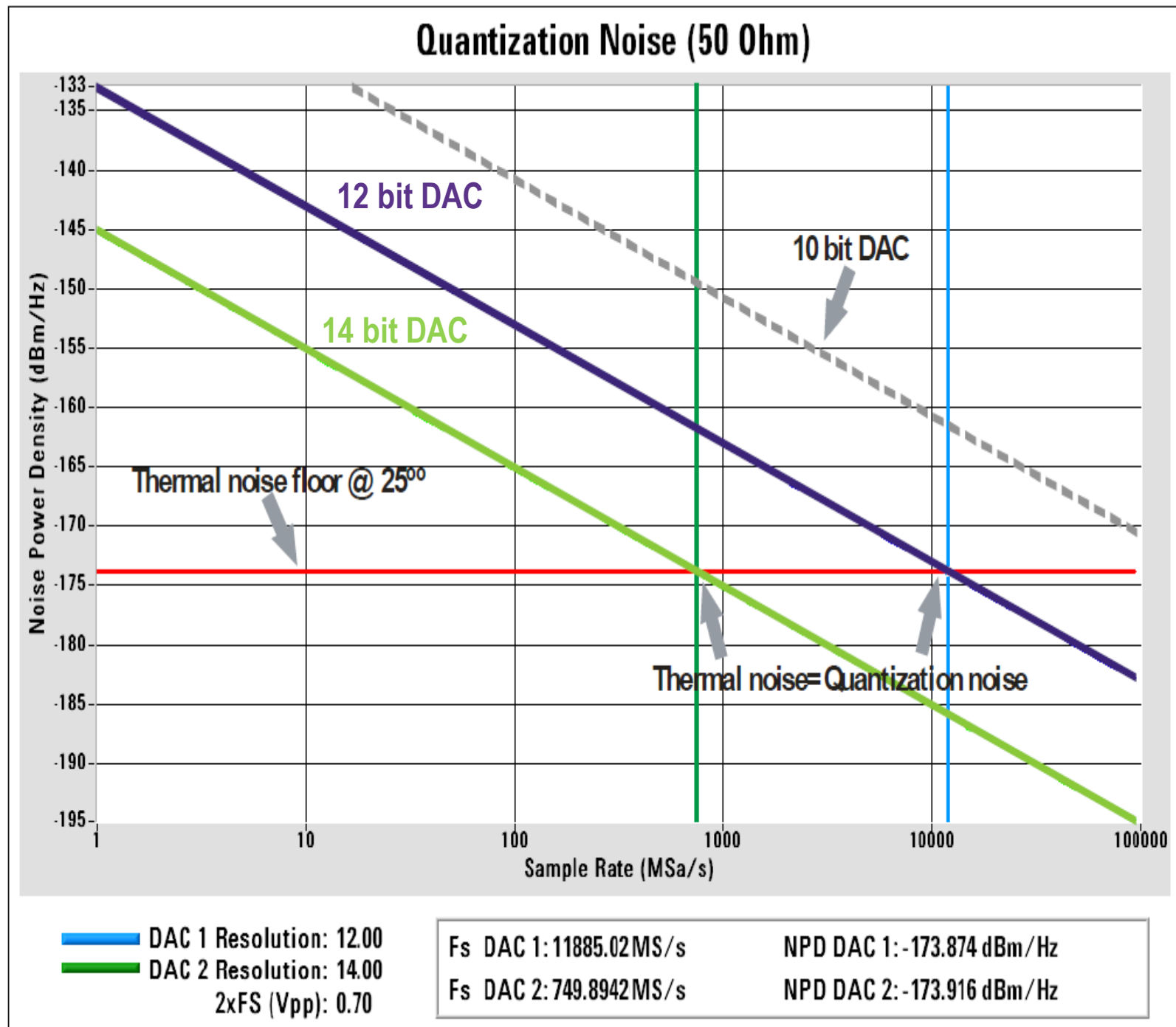
- A N-bit ADC has quantisation steps of  $n = 2^N$  .
- The Quantisation Error  $q$  of a signal with an Amplitude  $A$  is calculated using  $q = \frac{A}{2^N - 1}$  .

### Example for a signal with an amplitude of 1V at full scale:

- ✓ Vertical resolution for an 8-bit DAC is  $n = 2^8 = 256$  steps.
- ✓ The quantisation Error is then calculated to  $q = \frac{1V}{2^8 - 1} = 3.9 \text{ mV}$
  
- ✓ Vertical resolution for an 10-bit DAC is  $n = 2^{10} = 1024$  steps.
- ✓ The quantisation Error is then calculated to  $q = \frac{1V}{2^{10} - 1} = 0.98 \text{ mV}$
- ✓ Often  $q$  is also called „quantisation noise“ which can be analyzed easily within the frequency domain.

# Quantization Noise Power Density vs. Sample Rate

- In the Keysight M8190A AWG, with 14 bits resolution at 8GSa/s and 12 bits at 12 GSa/s, **quantization noise is negligible** in front of other noise sources, however, it may not stand true for a 10 bit instrument running at the same speed.

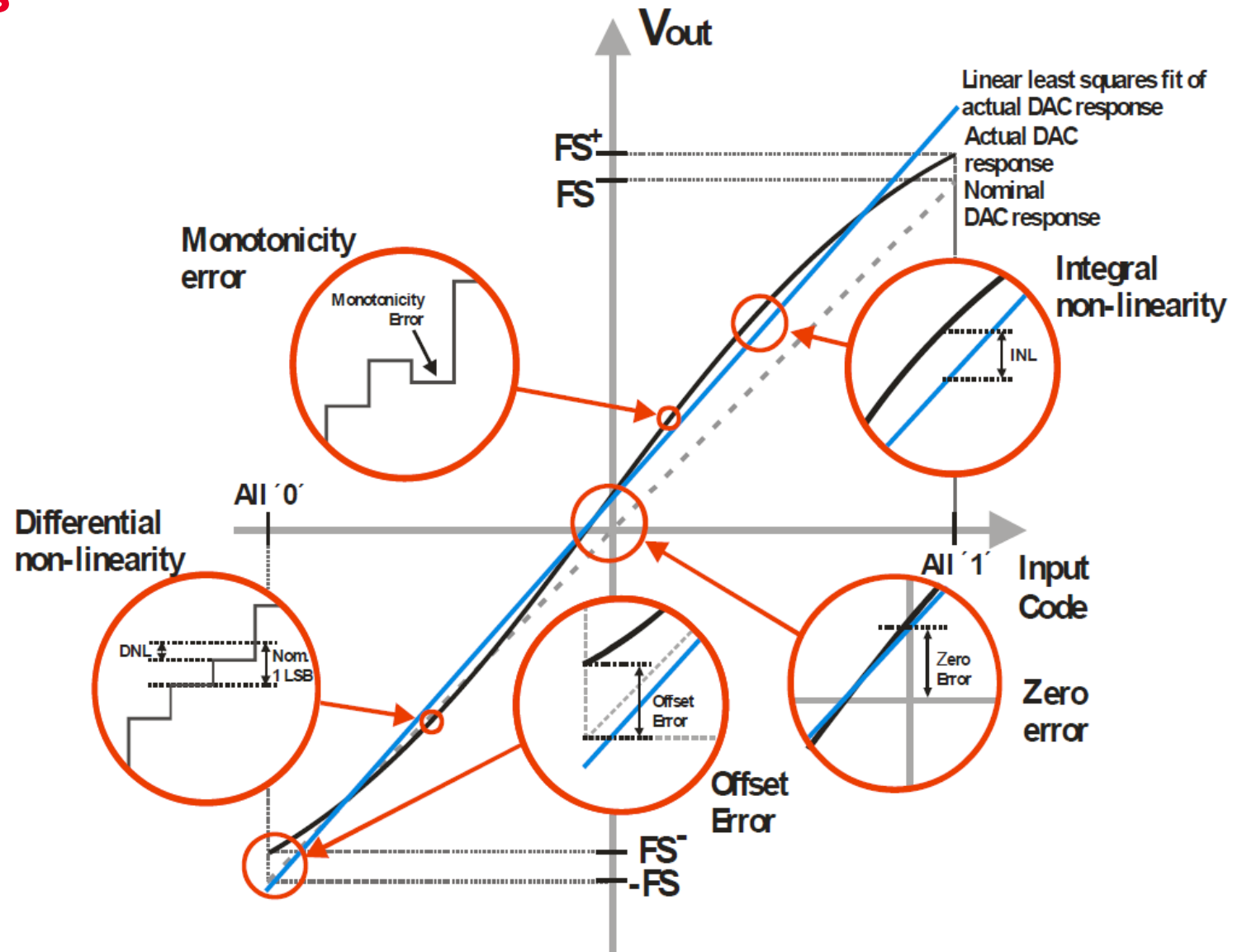




# AWG Basics: Non-Linearities

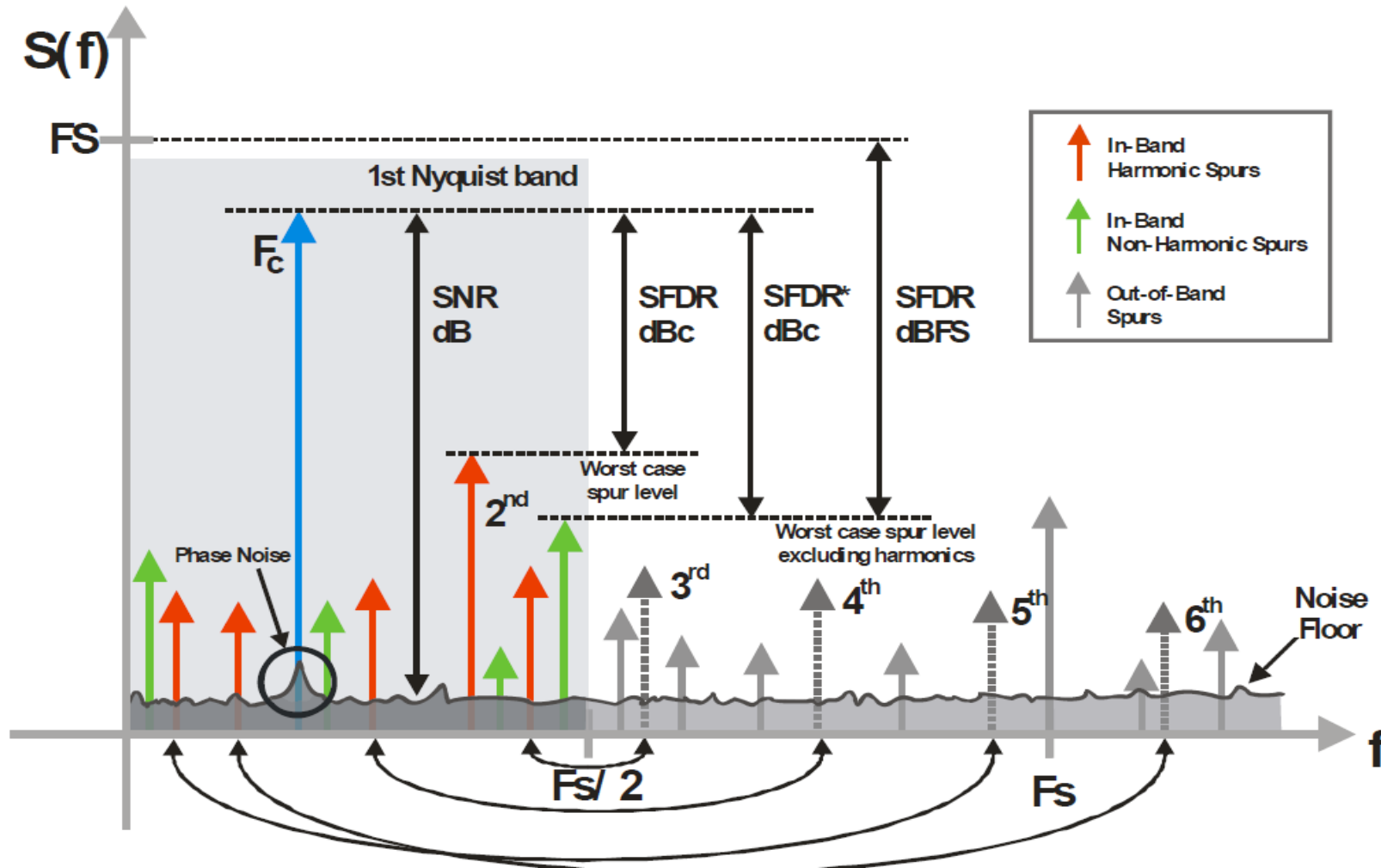
## Theory of AWG Operation

- Real world DACs are not perfect and transfer function deviates from the ideal response.
- Linear component will not result in harmonic or inter-modulation distortion in the output waveform as opposed to the non-linear components.



# AWG Basics: Non-Linearities

Frequency Domain



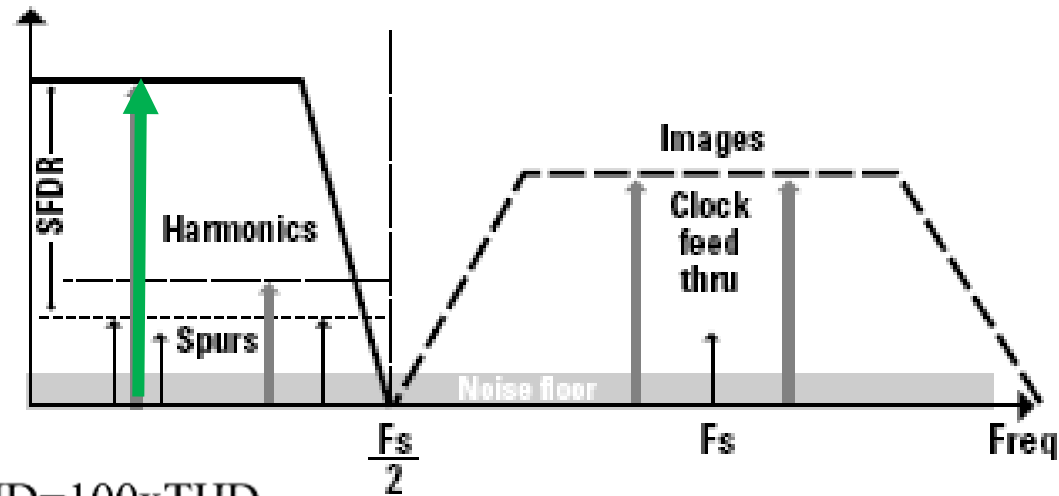
$$THD_{\%} = (\sum H_n^2)^{1/2} / S \times 100\%, \quad n=2 \dots N, \quad H_n \text{ and } S \text{ are rms values}$$

# Effective Number of Bits or ENOB

Impairments limit AWG Dynamic Range !

SFDR = Spurious Free Dynamic Range  
SINAD = ratio of the total signal power level (Signal + Noise + Distortion) to unwanted signal power (Noise + Distortion)

Images  
 Sample Clock Feedthru  
 Harmonics  
 Spurious  
 Noise Floor



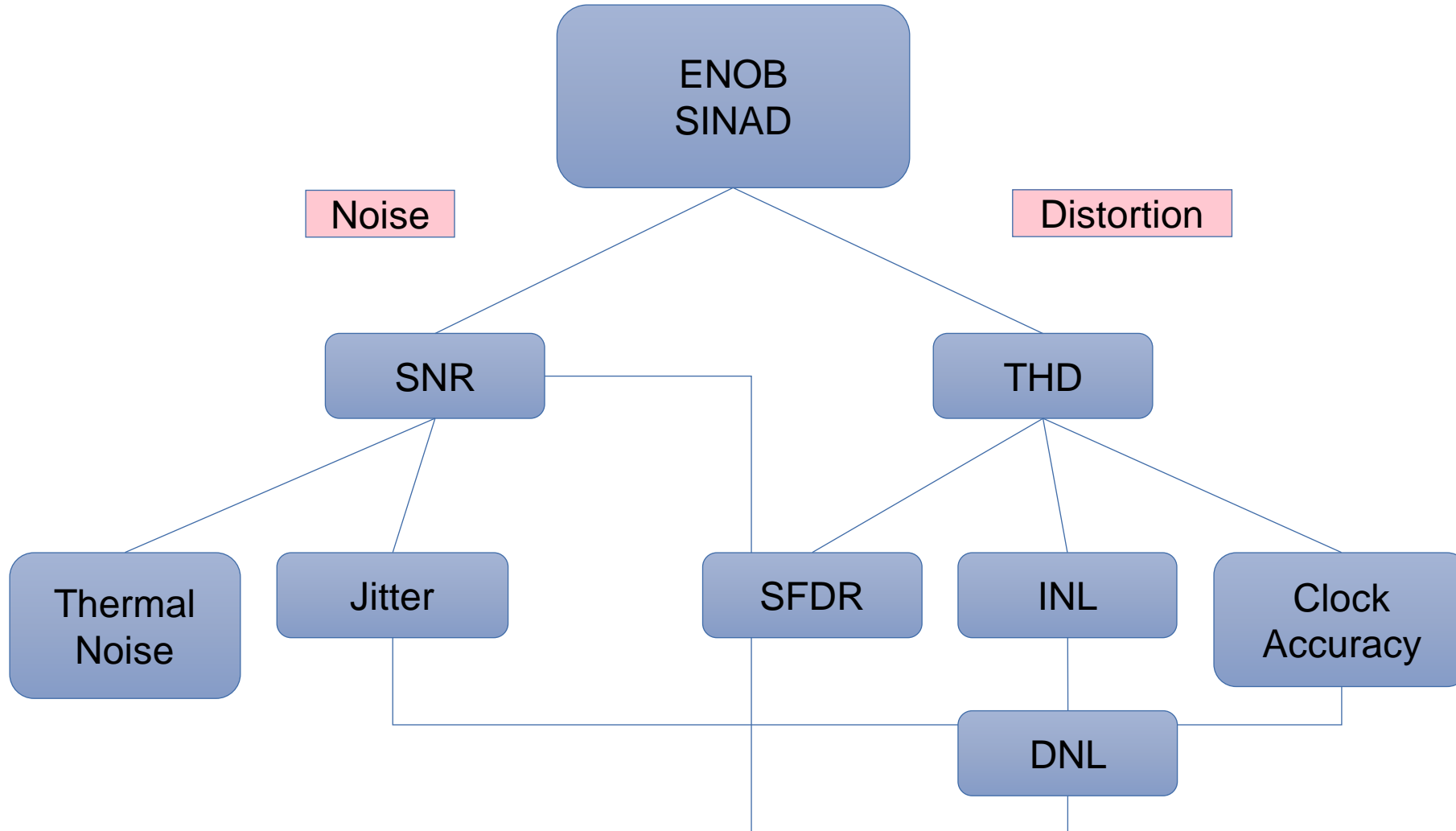
$$SINAD_{dBc} = 10\log_{10}(10^{-SNR/10} + 10^{-THD/10}), \quad THD = 100 \times THD\%$$

$$ENOB = \left( \frac{SINAD - 10\log(3|2)}{20\log 2} \right) = \frac{SINAD - 1.76}{6.02}$$

determines dynamic range  
 Every Bit in DAC doubles the voltage resolution  
 ~ 6 dB per bit

# Effective Number of Bits *or* ENOB

What Sources are Included in the Calculation?



## Effective Number of Bits *or* ENOB

What does ENOB ignore?

- Testing is done with a sine wave  
(which has no harmonics)
- Effective bits ***neglect*** these sources of error:
  - Amplitude Flatness
  - Phase Linearity
  - Gain Accuracy
  - Offset Accuracy

# High-Precision AWG Example

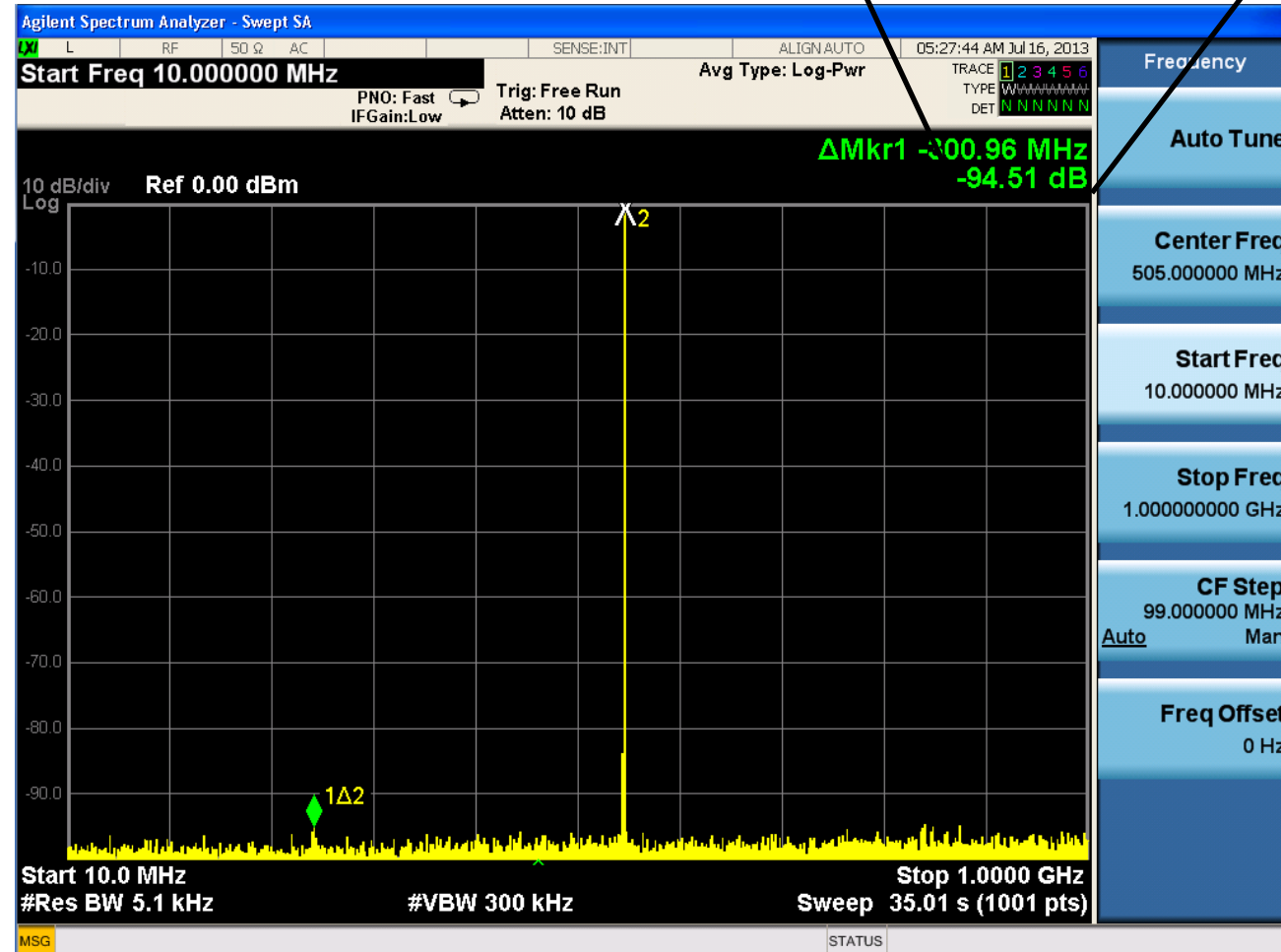
CW Signal

Single tone  
555 MHz

$F_s = 7.2$  GHz

Spurs:  $< -90$  dBc

(in the range  
0 to 1 GHz)



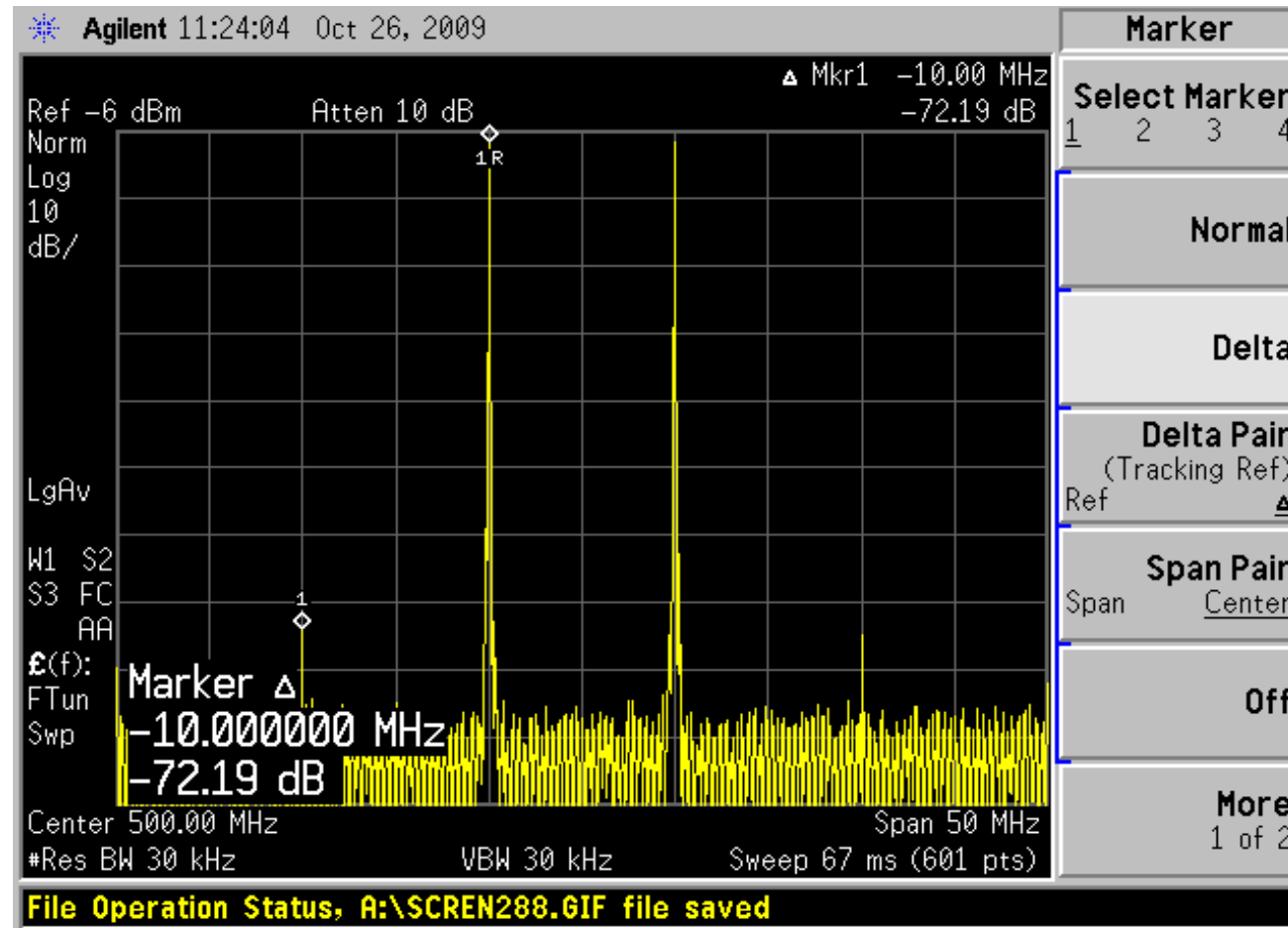
# High-Precision AWG Example:

## Two-Tone Signal

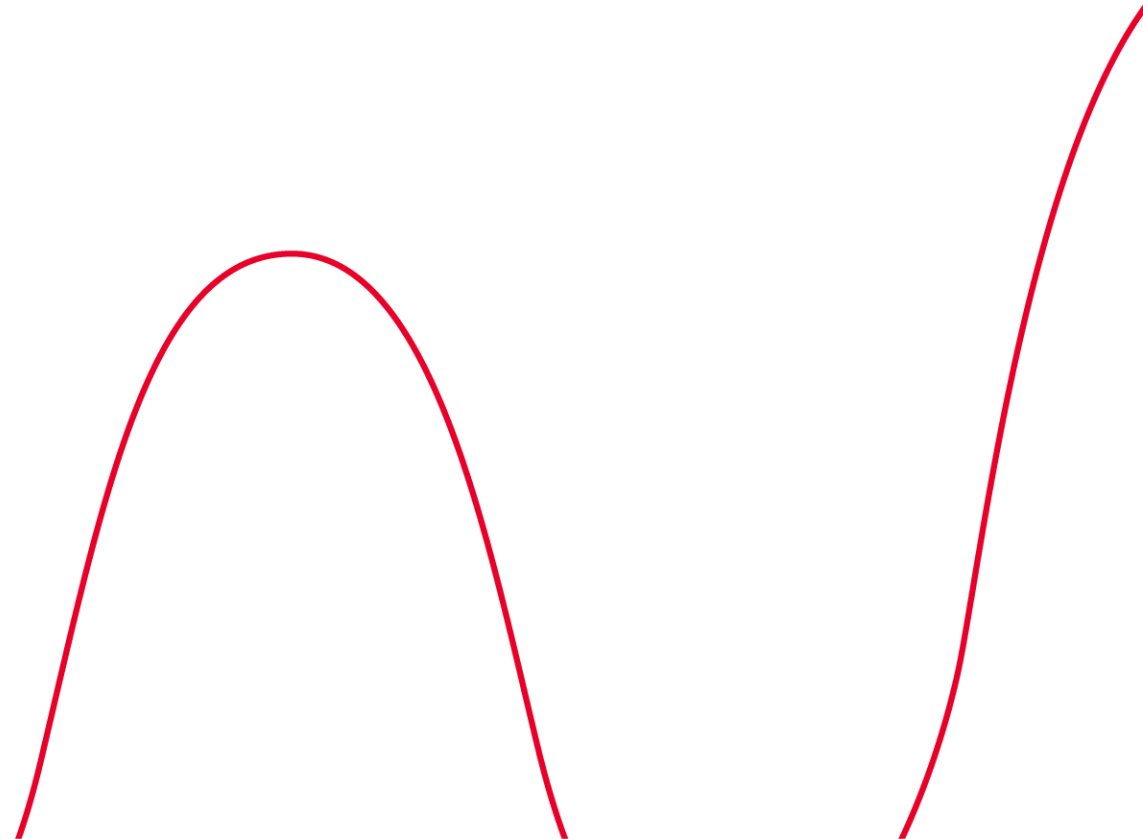
Two-tone signal  
Center 500 MHz  
Distance 10 MHz

$F_s = 7.2 \text{ GHz}$

IMD: -72 dBc



**It's time for a demo...**





# Agenda

## AWG Workshop

Arbitrary Waveform Generation Fundamentals

Frequency Response Correction

In-System Calibration

Conclusion & Summary

# Corrections (1)

## Types

- Channel-specific frequency and phase response
  - Frequency and phase response has been characterized individually per channel and for different output amplitudes. Range is 32 GHz in 10 MHz steps.
  - Reference is the output connector.
  - Stored in the calibration flash.
- Standard cable corrections based on a typical high-quality, high-bandwidth 0.85m microwave cable (Huber+Suhner type M8041-61616)
- File with correction factors. Format is compatible with adaptive equalizer files exported in CSV format from the Keysight 89600 VSA.



# Corrections (2)

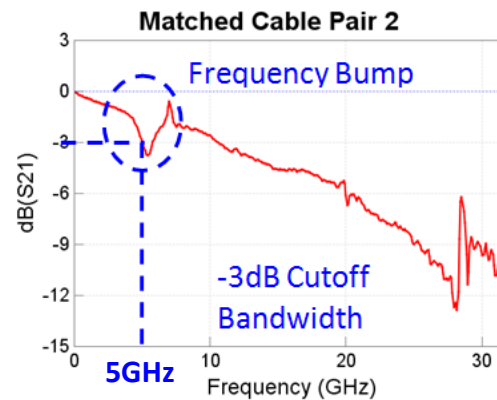
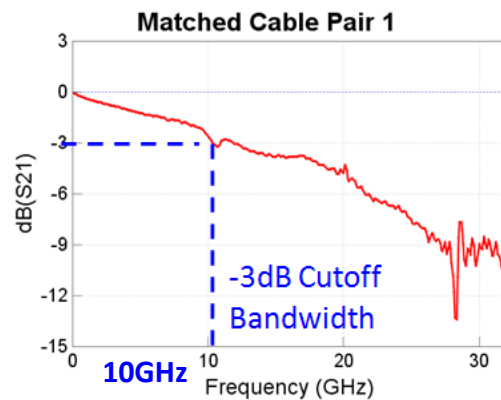
## File Format

```
// MyCorrectionFile
InputBlockSize, 1024 // number of correction factors
XStart, 1.0E+09      // first factor is at 1 GHz
XDelta, 1.0E+06     // step size is 1 MHz
YUnit, lin          // lin: linear relative amplitude, dB: logarithmic relative amplitude
Y
0.987, -0.2343      // amplitude, phase in radians
0.995, 0.5674
...
...
1.269, -0.765
```

# RF Cables Performance Matters!

## Element Models affect correction Results

- Good element models (S-parameter) are critical to accurate de-embedding/correction results.
- One of the biggest pitfall of de-embedding is using questionable element models.
  - Always confirm the S-parameter represents your circuit components.
  - Don't assume model stays the same for similar parts. Performance can vary even they are the same component and circuit.
  - Verify the performance with Network Analyzer whenever possible.



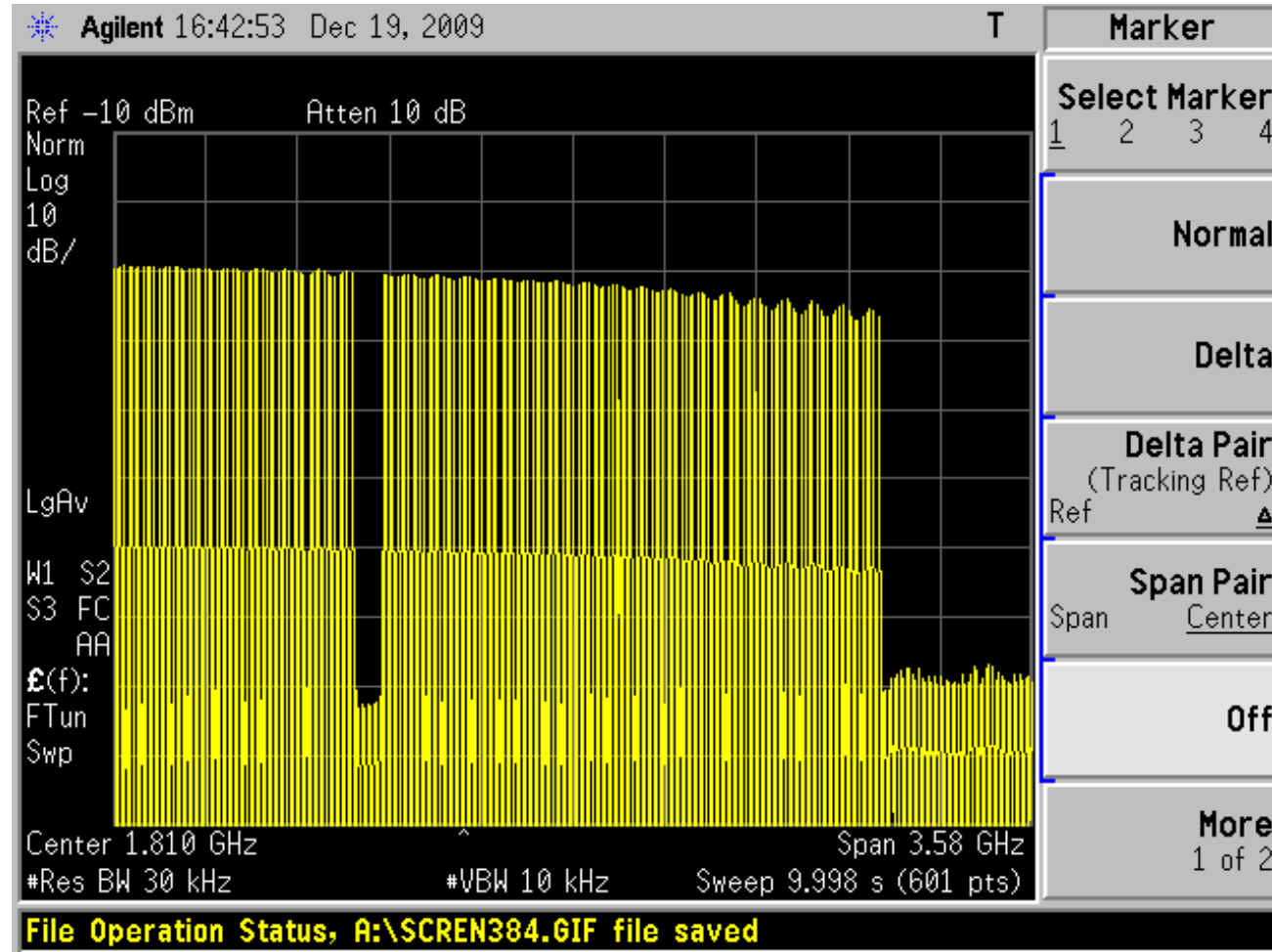
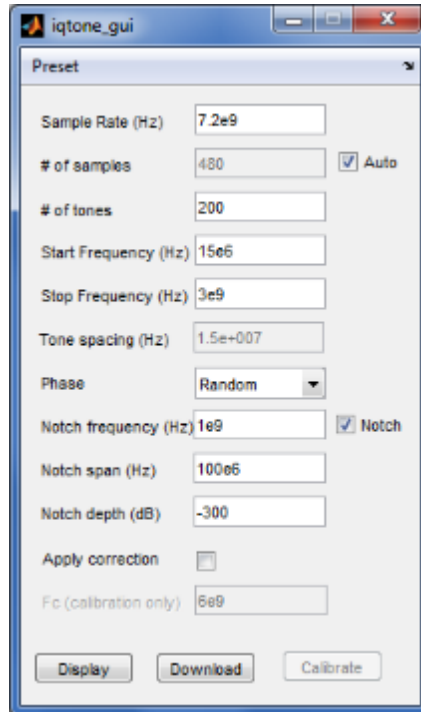
Responses of a matched cable pair. They are not exactly matched!

# Amplitude Correction

## Multi-Tone Signal

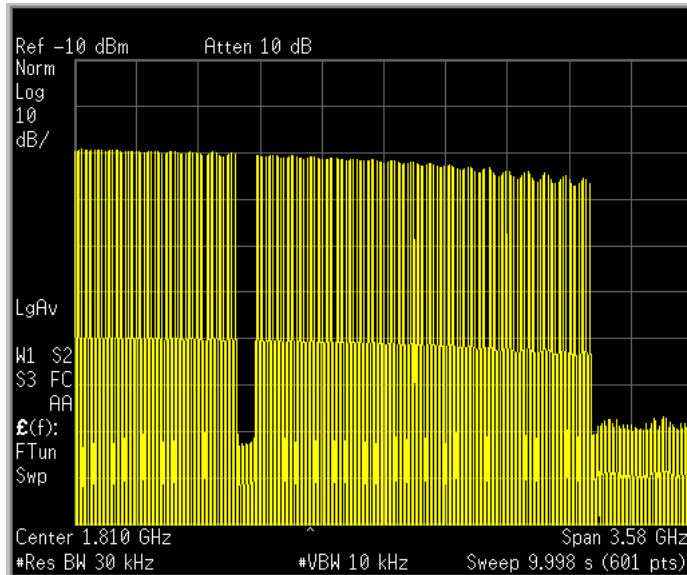
Multi-tone signal with 200 tones,  
3 GHz bandwidth

$F_s = 7.2$  GHz  
without amplitude correction

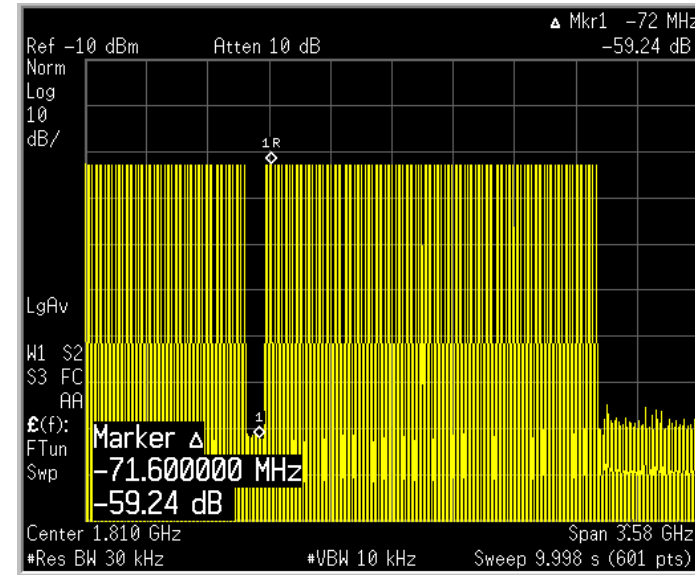
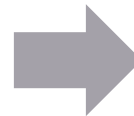


# Amplitude Correction

Frequency and phase response correction



Original Signal

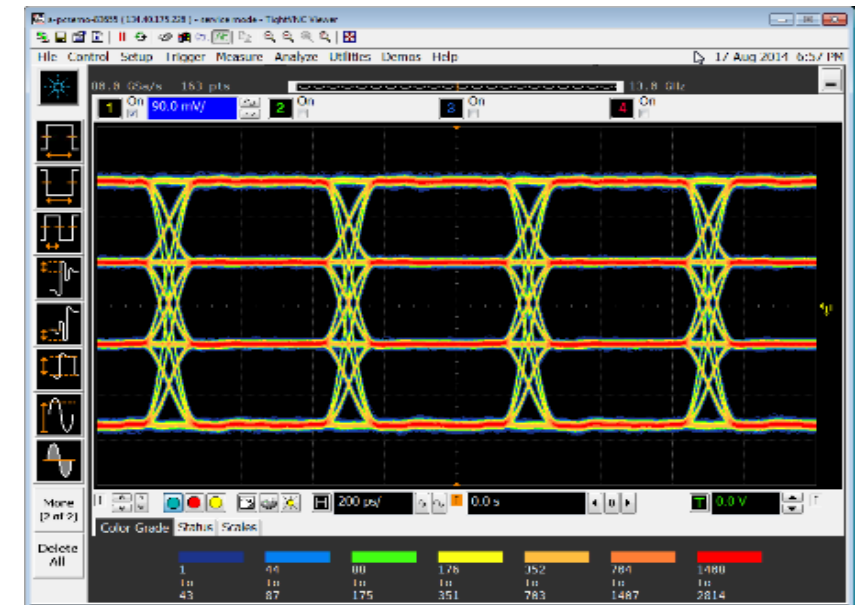
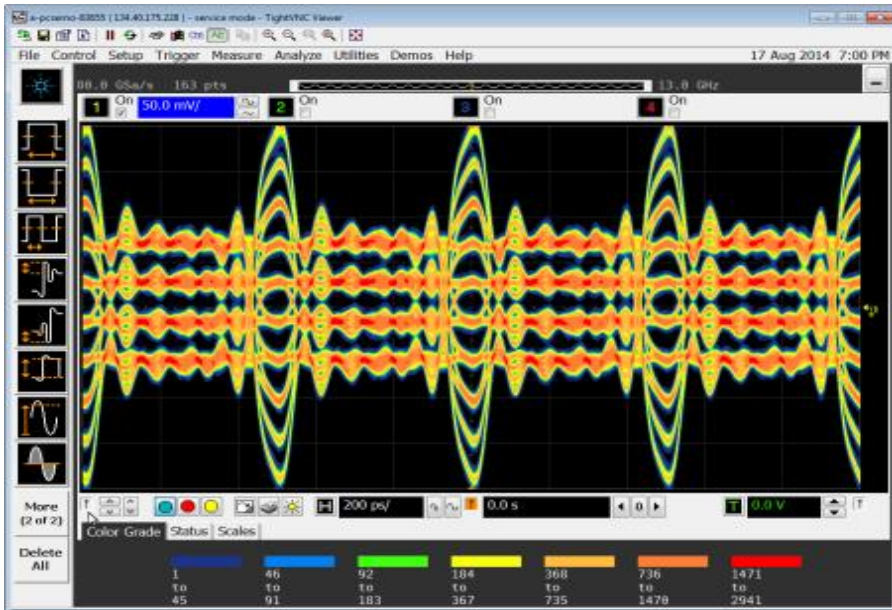
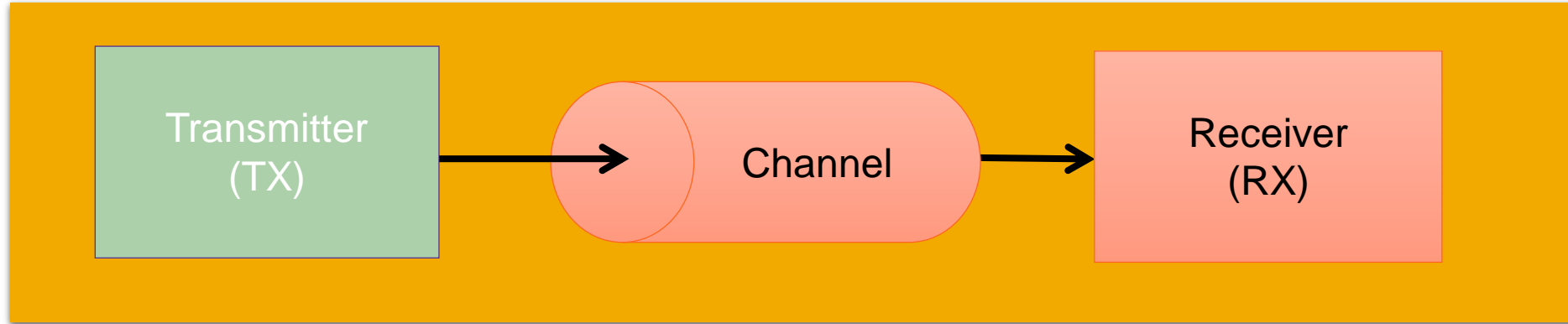


With digital pre-distortion

Depending on the bandwidth and carrier frequency, the flatness can be calibrated to around 0.1 dB flatness

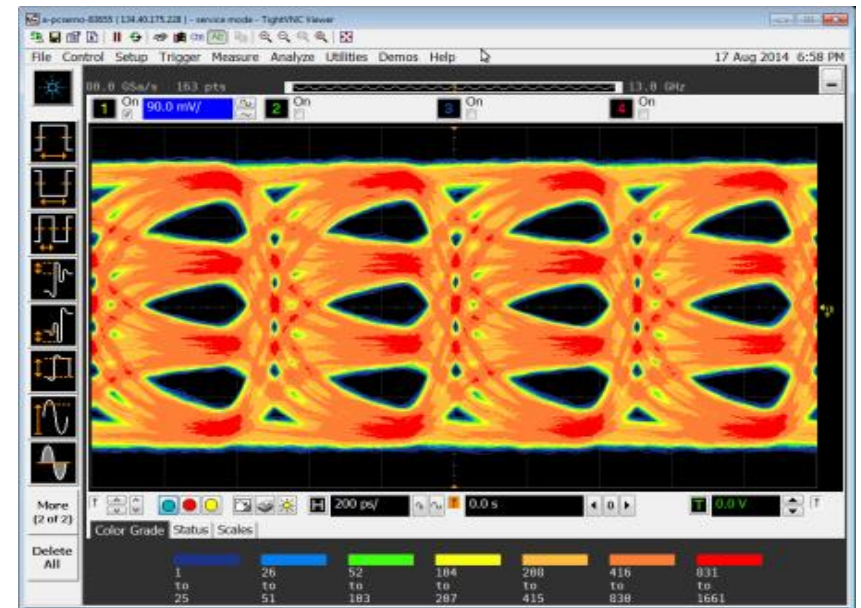
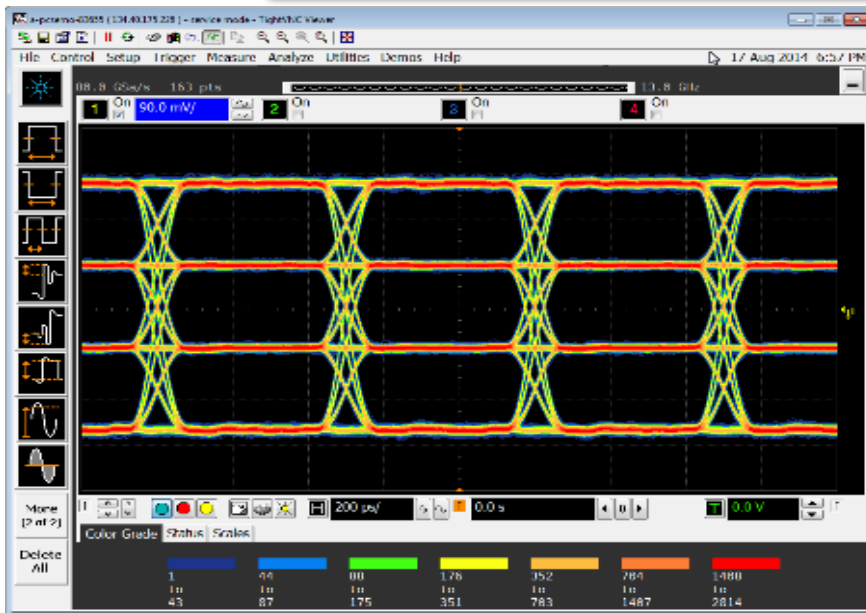
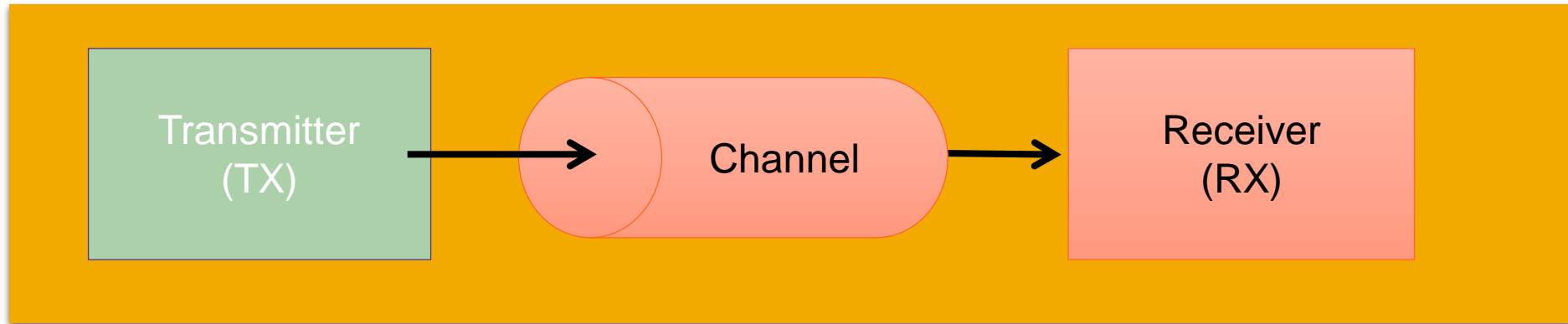
# Channel De-embedding

Using S-Parameters or measured in-system



# Channel Embedding

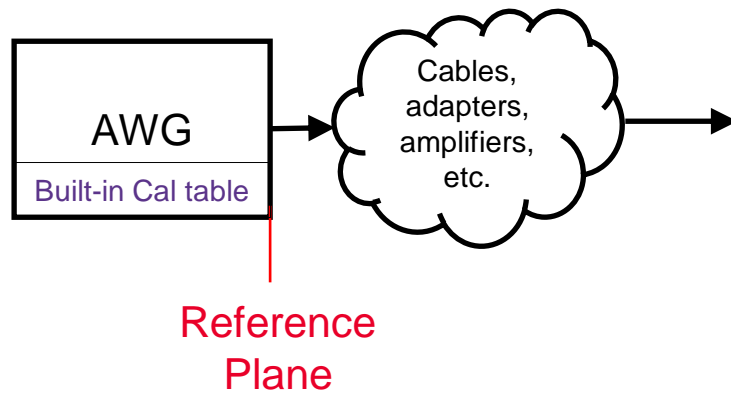
Using S-Parameters or measured in-system



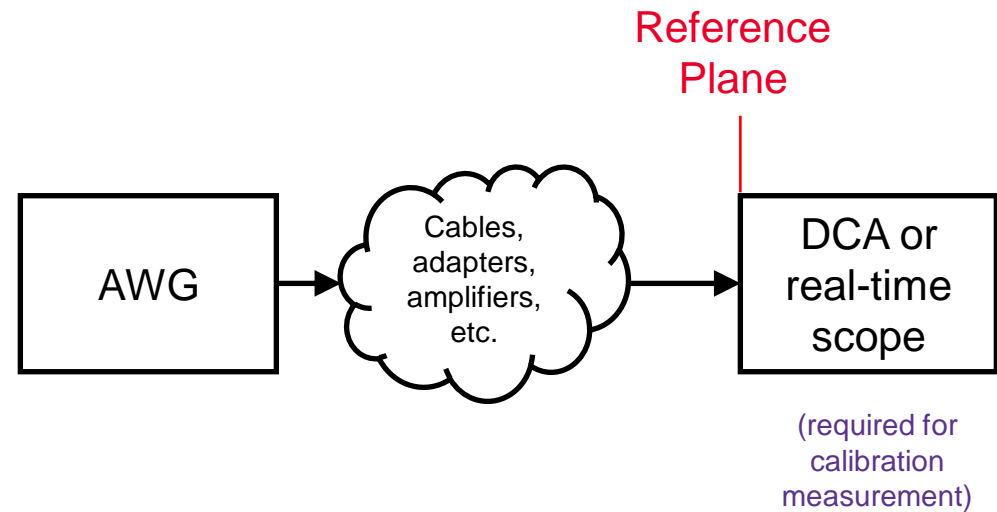


# Where is the Reference Plane?

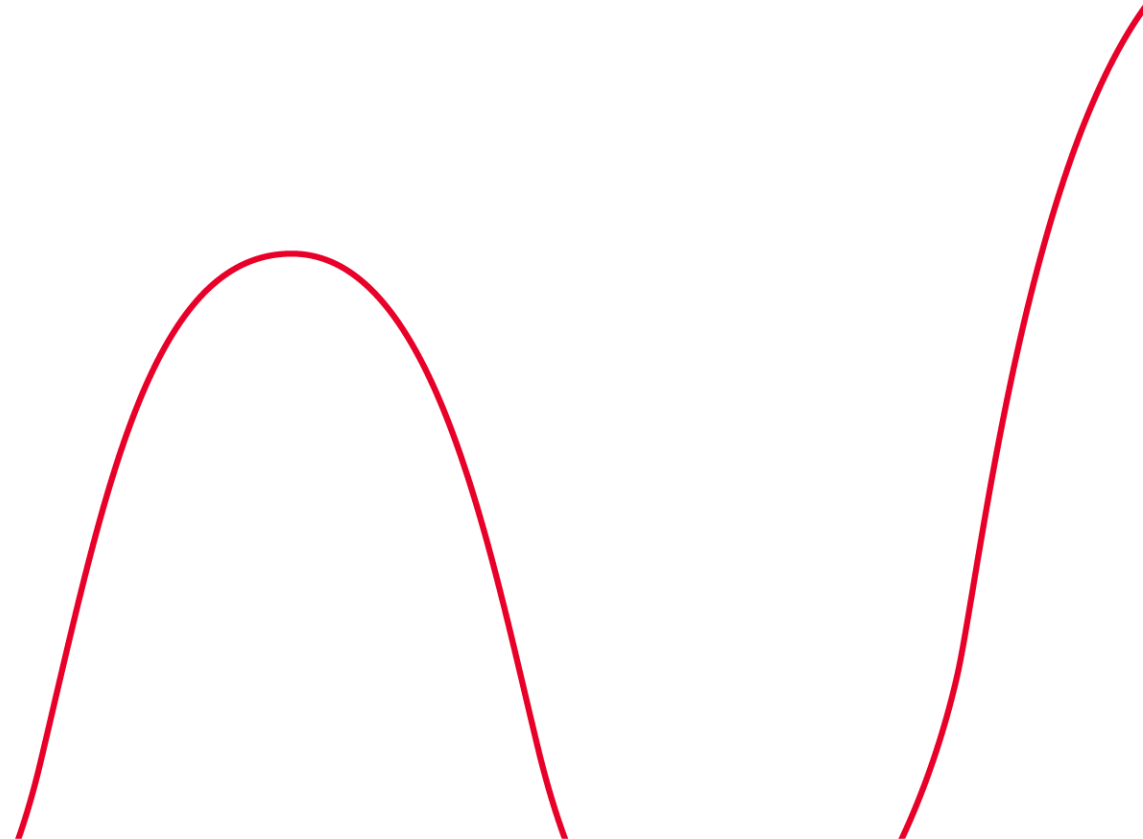
## AWG factory calibration



## The desired reference plane location



**It's time for a demo...**



# Agenda

## AWG Workshop

Arbitrary Waveform Generation Fundamentals

Frequency Response Correction

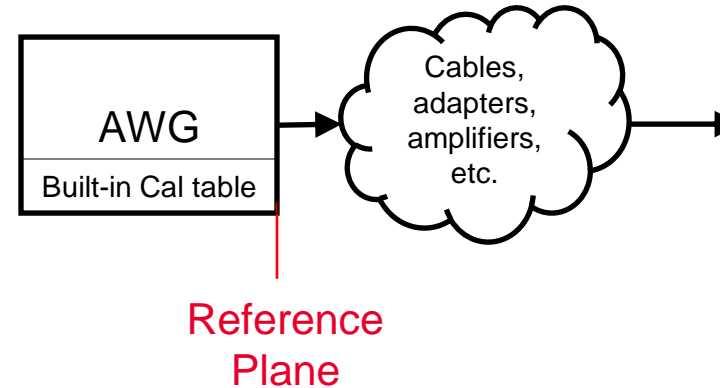
**In-System Calibration**

Conclusion & Summary

# Built-in Vs. In-system Frequency/Phase Response Calibration

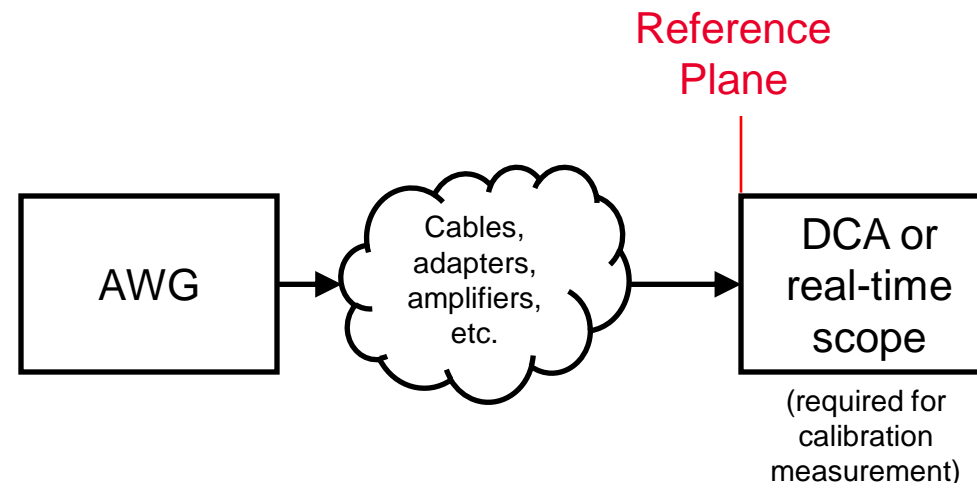
- **Built-in correction**

- No calibration measurement required
- Additional circuits can be deembedded using S-parameters



- **In-system calibration**

- More accurate
- One-time calibration measurement required
- Takes external circuitry into account
- Setup does not need to be modified



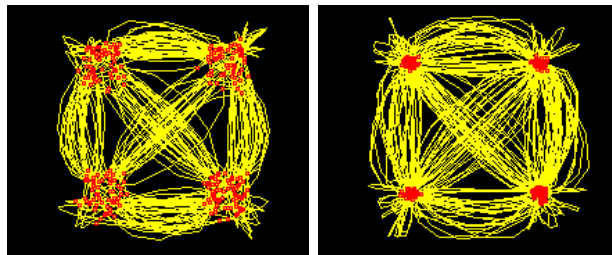
# Accurate and Repeatable Test Results

Out-of-the-box calibration to ensure clean signal at the front connector

In-situ calibration – extend clean signal to the receiver test point

- S-Parameters of channel are embedded or de-embedded
- Frequency/phase response is measured in-system and then de-embedded

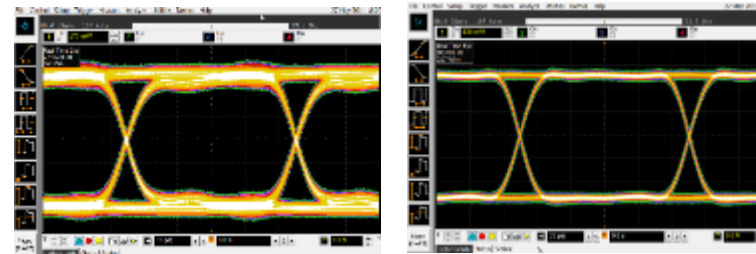
QPSK, 32 Gbaud



Without correction

With correction

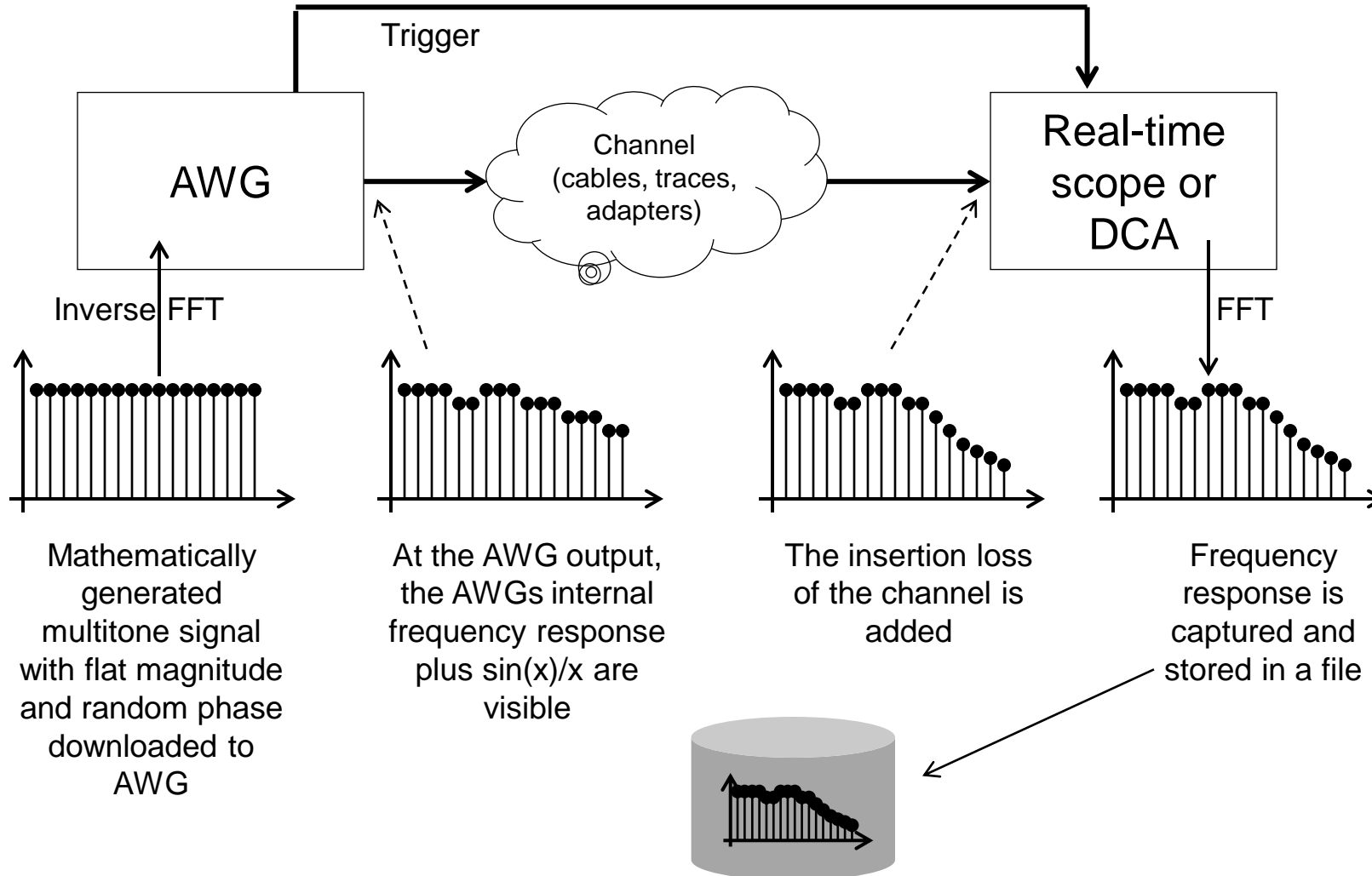
PRBS 6 Gbit/s



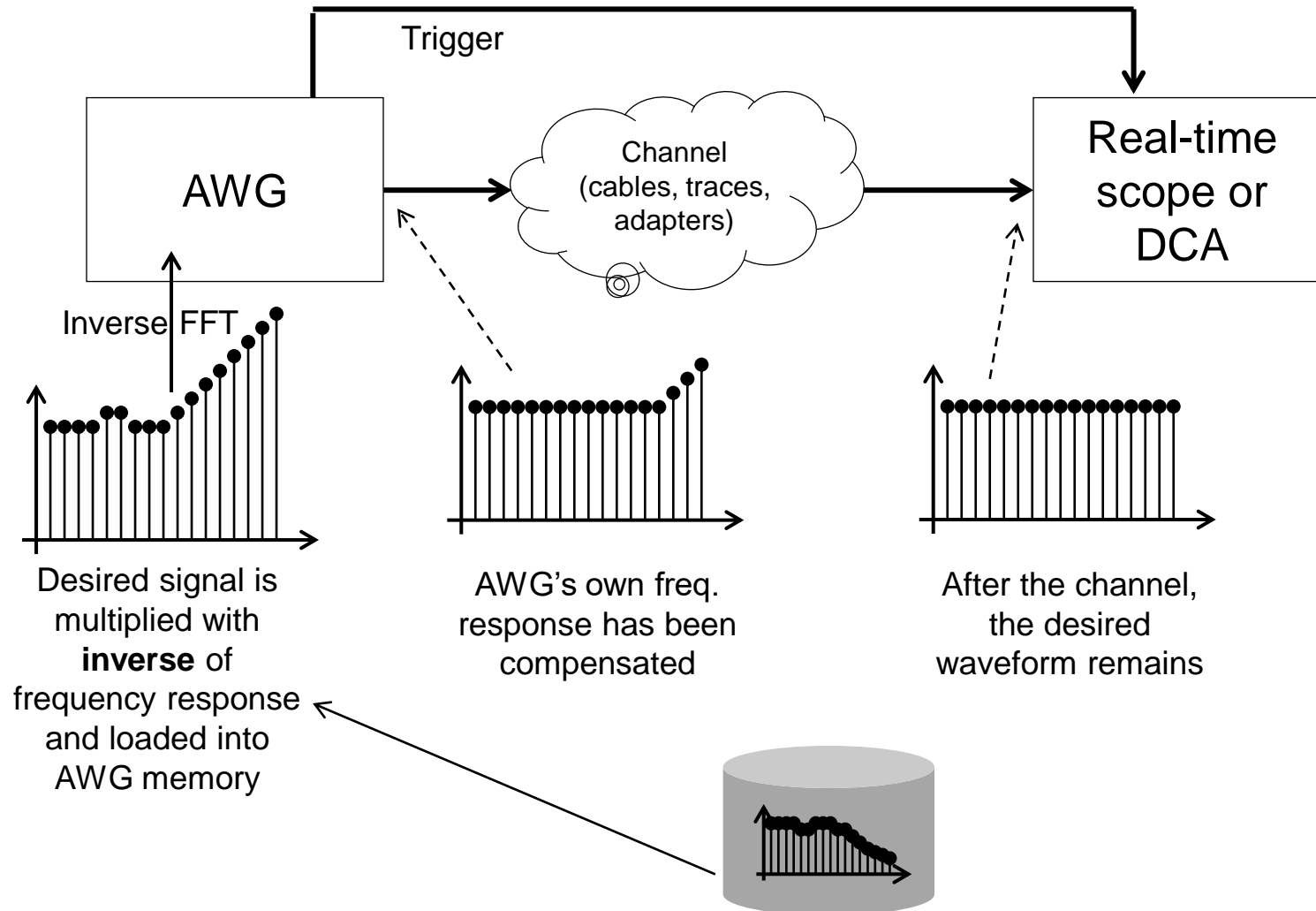
Without correction

With correction

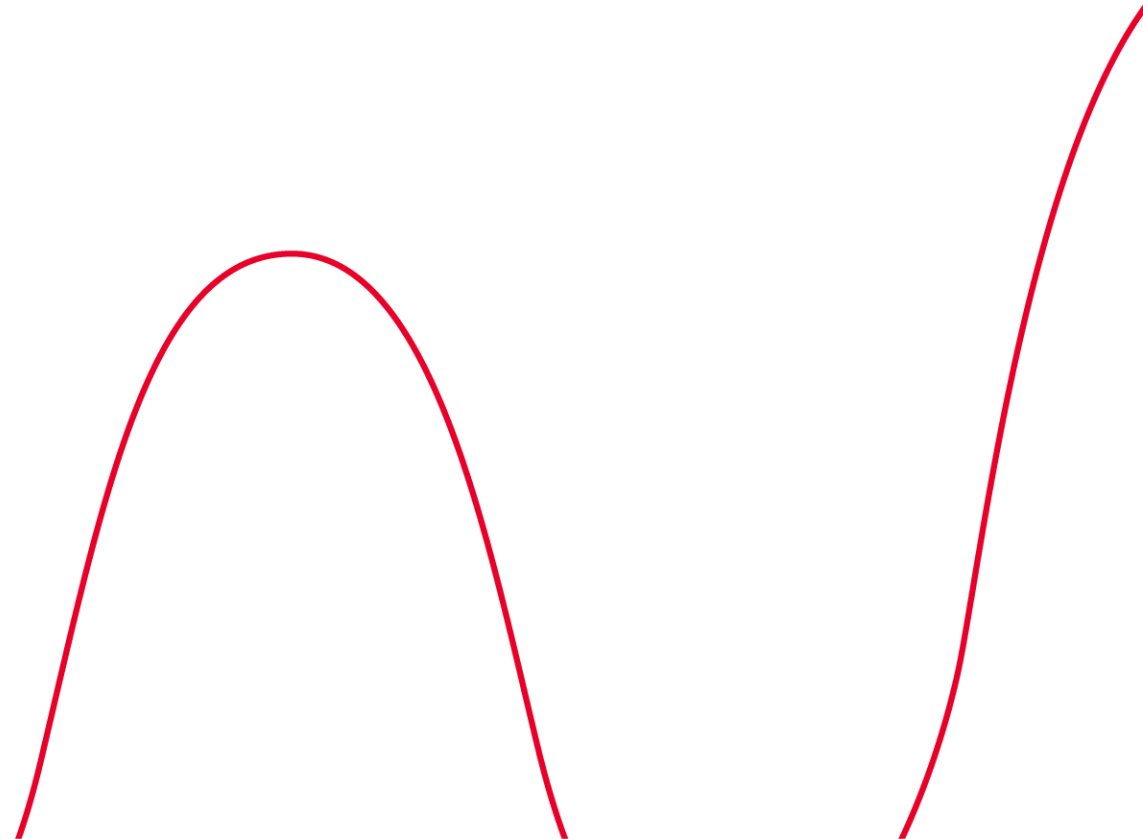
# How Does In-system Calibration Work?



# How Is In-system Calibration Applied?



**It's time for a demo...**





# Agenda

## AWG Workshop

Arbitrary Waveform Generation Fundamentals

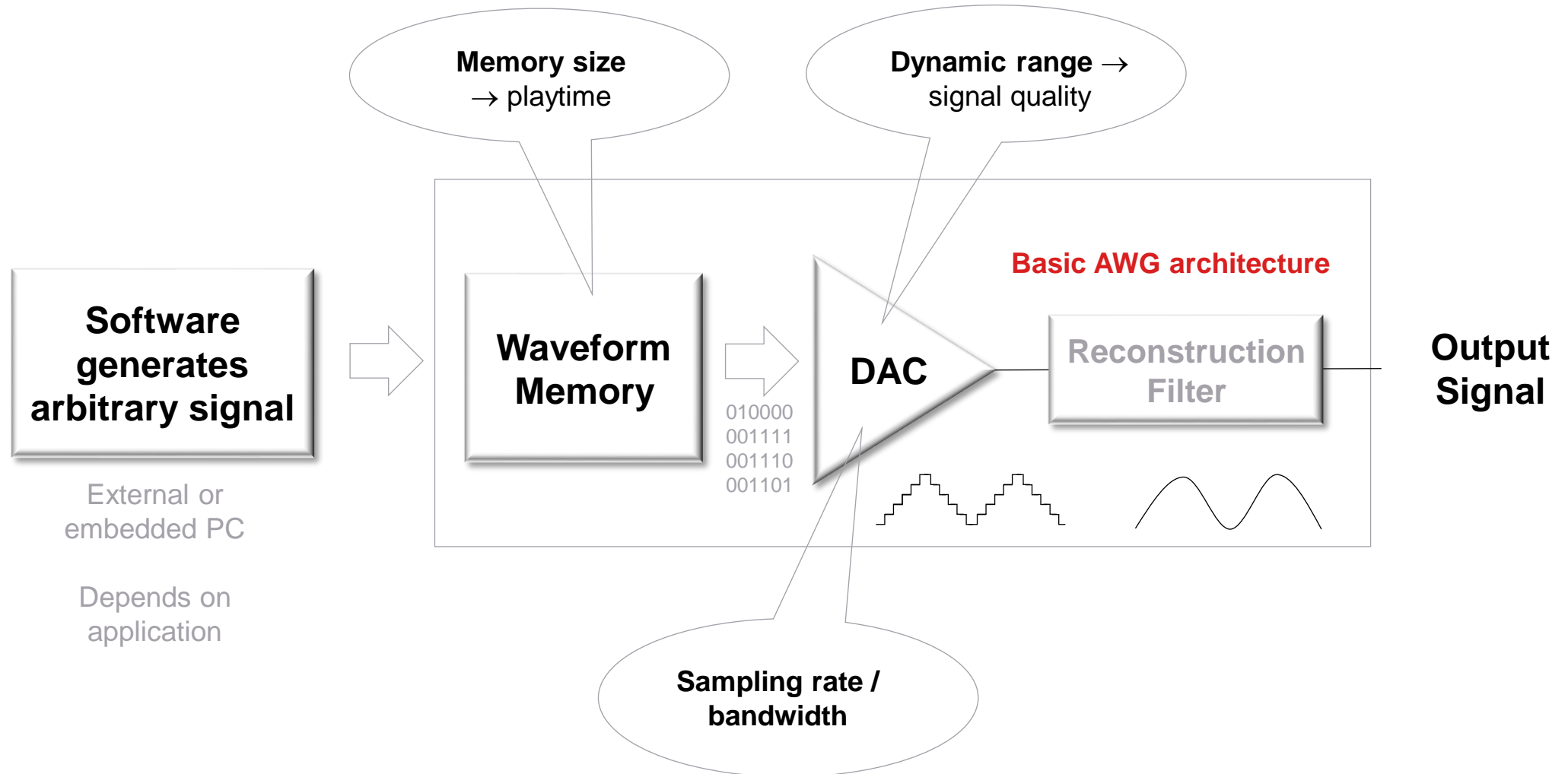
Frequency Response Correction

In-System Calibration

Conclusion & Summary

# Review of the Previous

## Key Blocks and Specifications



## Short Quiz

### Question 1

**Which of the following is not true about AWGs?**

- a) AWGs provide more flexibility in generating pulse shapes and pre-distortion than pulse generators typically do
- b) AWGs typically can't generate multiple carriers
- c) AWGs typically have wider modulation bandwidth vs signal generators
- d) AWGs typically have less dynamic range than signal generators

## Short Quiz

### Question 1

**Which of the following is not true about AWGs?**

- a) AWGs provide more flexibility in generating pulse shapes and pre-distortion than pulse generators typically do
- b) AWGs typically can't generate multiple carriers**
- c) AWGs typically have wider modulation bandwidth vs signal generators
- d) AWGs typically have less dynamic range than signal generators

## Short Quiz

### Question 2

**The sampling frequency for an AWG should be at least twice the highest frequency contained in the signal.**

- a) True
- b) False
- c) Other...

## Short Quiz

### Question 2

The sampling frequency for an AWG should be at least twice the highest frequency contained in the signal.

- a) True
- b) False
- c) **Other...**

## Short Quiz

### Question 3

**Using s-parameters to simulate the effects of running a signal through additional channels is called**

- a) De-embedding
- b) Embedding

## Short Quiz

### Question 3

Using s-parameters to simulate the effects of running a signal through additional channels is called

- a) De-embedding
- b) **Embedding**



## Short Quiz

### Question 4

**An AWG's playback time can be improved by**

- a) Decreasing its sample rate
- b) Using other techniques such as sequencing
- c) Both of these could improve an AWG's playback time

## Short Quiz

### Question 4

**An AWG's playback time can be improved by**

- a) Decreasing its sample rate
- b) Using other techniques such as sequencing
- c) Both of these could improve an AWG's playback time**

## Short Quiz

### Question 5

**An AWG's reconstruction filter limits the bandwidth and “smooths out” the waveform.**

- a) True
- b) False

## Short Quiz

### Question 5

**An AWG's reconstruction filter limits the bandwidth and “smooths out” the waveform.**

- a) **True**
- b) False

**Thank you**

