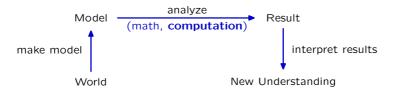
6.3000: Signal Processing

Sampling and Aliasing

Importance of Discrete Representations

Our goal is to develop **signal processing** tools to model interesting aspects of the world, to analyze the model, and to interpret the results.



The increasing power and decreasing cost of computation makes the use of computation increasingly attractive.

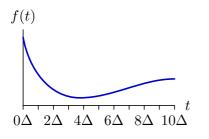
However, many important signals are naturally described with continuous functions, that must be **sampled** in order to be analyzed computationally.

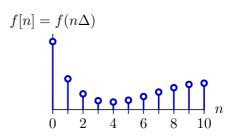
Today: understand relations between **continuous** and **sampled** signals.

Sampling

Sampling refers to the process by which a continuous-time signal f(t) is converted to a discrete-time signal f[n].

We use parentheses to denote functions of continuous domain (e.g., f(t)) and square brackets to denote functions of discrete domain (e.g., f[n]).





$$\Delta=$$
 sampling interval $f_s=rac{1}{\Delta}=$ sampling frequency

How does sampling affect the information contained in a signal?

Sampling Music

$$f_s = \frac{1}{\Delta}$$

- $f_s = 44.1 \text{ kHz}$ • $f_s = 22 \text{ kHz}$
- $f_s = 11 \text{ kHz}$
- $\bullet~f_s=5.5~\mathrm{kHz}$
- $f_s = 2.8 \text{ kHz}$

J.S. Bach, Sonata No. 1 in G minor Mvmt. IV. Presto Nathan Milstein, violin

Sampling Images



original: 2048×1536

Sampling Images



downsampled: 1024×768

Sampling Images



downsampled: 512×384

Sampling Images



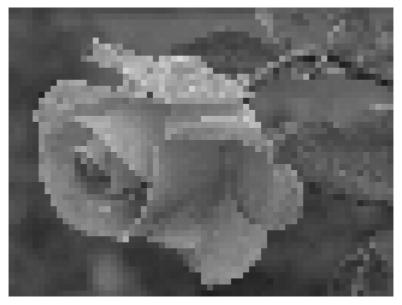
downsampled: 256×192

Sampling Images



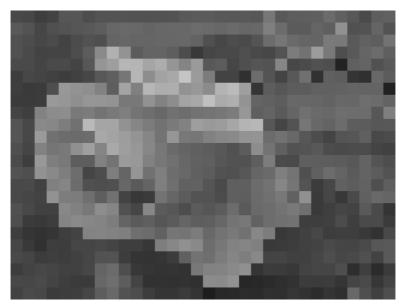
downsampled: 128×96

Sampling Images



downsampled: 64×48

Sampling Images

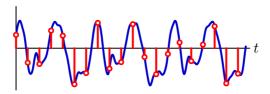


downsampled: 32×24

We would like to sample in a way that preserves information.

However, information is often ${\color{red} \textbf{lost}}$ in the sampling process.

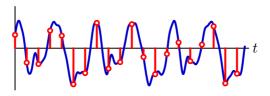
Example: samples (red) provide no information about intervening values.



We would like to sample in a way that preserves **information**.

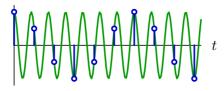
However, information is generally **lost** in the sampling process.

Example: samples (red) provide no information about intervening values.



Furthermore, information that is retained by sampling can be misleading.

Example: samples can suggest patterns not contained in the original.

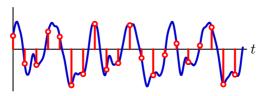


Samples (blue) of the original high-frequency signal (green)

We would like to sample in a way that preserves information.

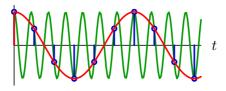
However, information is generally **lost** in the sampling process.

Example: samples (red) provide no information about intervening values.



Furthermore, information that is retained by sampling can be misleading.

Example: samples can suggest patterns not contained in the original.



Samples (blue) suggest an input that is much lower in frequency (red) than the original signal (green).

Our goal is to understand sampling so that we can mitigate its effects on the information contained in the signals we process.

First, consider sampling a cosine function with fixed frequency $\omega=2\pi.$

$$x(t) = \cos(2\pi t)$$

$$\Delta = 0.1$$

$$x[n] = \cos(2\pi \Delta n) = \cos(2\pi 0.1 n)$$

$$n, t/\Delta$$

First, consider sampling a cosine function with fixed frequency $\omega=2\pi.$

$$x(t) = \cos(2\pi t)$$

$$\Delta = 0.2$$

$$x[n] = \cos(2\pi \Delta n) = \cos(2\pi 0.2 n)$$

$$n, t/\Delta$$

First, consider sampling a cosine function with fixed frequency $\omega=2\pi.$

$$x(t) = \cos(2\pi t)$$

$$\Delta = 0.3$$

$$x[n] = \cos(2\pi \Delta n) = \cos(2\pi 0.3 n)$$

$$n, t/\Delta$$

First, consider sampling a cosine function with fixed frequency $\omega=2\pi.$

$$x(t) = \cos(2\pi t)$$

$$\Delta = 0.4$$

$$x[n] = \cos(2\pi \Delta n) = \cos(2\pi 0.4 n)$$

$$n, t/\Delta$$

First, consider sampling a cosine function with fixed frequency $\omega=2\pi.$

$$x(t) = \cos(2\pi t)$$

$$\Delta = 0.5$$

$$x[n] = \cos(2\pi \Delta n) = \cos(2\pi 0.5 n)$$

$$n, t/\Delta$$

First, consider sampling a cosine function with fixed frequency $\omega=2\pi.$

$$x(t) = \cos(2\pi t)$$

$$\Delta = 0.6$$

$$x[n] = \cos(2\pi \Delta n) = \cos(2\pi 0.6 n)$$

$$n, t/\Delta$$

First, consider sampling a cosine function with fixed frequency $\omega=2\pi.$

$$x(t) = \cos(2\pi t)$$

$$\Delta = 0.6$$

$$x[n] = \cos(2\pi \Delta n) = \cos(2\pi 0.6 n) = \cos(2\pi 0.4 n)$$

$$n, t/\Delta$$

$$\cos(2\pi \, 0.6 \, n) = \cos(-2\pi \, 0.6 \, n) = \cos(-2\pi \, 0.6 \, n + 2\pi n)$$
$$= \cos(2\pi (1 - 0.6) n) = \cos(2\pi \, 0.4 \, n)$$

First, consider sampling a cosine function with fixed frequency $\omega=2\pi.$

$$x(t) = \cos(2\pi t)$$

$$\Delta = 0.7$$

$$x[n] = \cos(2\pi \Delta n) = \cos(2\pi 0.7 n) = \cos(2\pi 0.3 n)$$

$$n, t/\Delta$$

$$\cos(2\pi \, 0.7 \, n) = \cos(-2\pi \, 0.7 \, n) = \cos(-2\pi \, 0.7 \, n + 2\pi n)$$
$$= \cos(2\pi (1 - 0.7) n) = \cos(2\pi \, 0.3 \, n)$$

First, consider sampling a cosine function with fixed frequency $\omega=2\pi.$

$$x(t) = \cos(2\pi t)$$

$$\Delta = 0.8$$

$$x[n] = \cos(2\pi \Delta n) = \cos(2\pi 0.8 n) = \cos(2\pi 0.2 n)$$

$$n, t/\Delta$$

$$\cos(2\pi \, 0.8 \, n) = \cos(-2\pi \, 0.8 \, n) = \cos(-2\pi \, 0.8 \, n + 2\pi n)$$
$$= \cos(2\pi (1 - 0.8) n) = \cos(2\pi \, 0.2 \, n)$$

First, consider sampling a cosine function with fixed frequency $\omega=2\pi.$

$$x(t) = \cos(2\pi t)$$

$$\Delta = 0.9$$

$$x[n] = \cos(2\pi \Delta n) = \cos(2\pi 0.9 n) = \cos(2\pi 0.1 n)$$

$$n, t/\Delta$$

$$\cos(2\pi \, 0.9 \, n) = \cos(-2\pi \, 0.9 \, n) = \cos(-2\pi \, 0.9 \, n + 2\pi n)$$
$$= \cos(2\pi (1 - 0.9) n) = \cos(2\pi \, 0.1 \, n)$$

First, consider sampling a cosine function with fixed frequency $\omega=2\pi.$

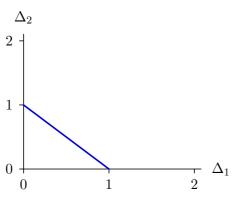
$$x(t) = \cos(2\pi t)$$

$$\Delta = 1.0$$

$$x[n] = \cos(2\pi \Delta n) = \cos(2\pi 1.0 n) = \cos(2\pi 0.0 n)$$

$$n, t/\Delta$$

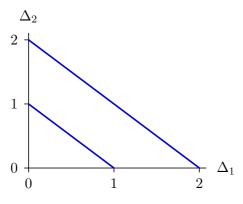
The same sequence of samples results when $x(t) = \cos(2\pi t)$ is sampled at intervals Δ_1 or Δ_2 if $\Delta_2 = 1 - \Delta_1$.



$$x[n] = \cos(2\pi\Delta_2 n) = \cos(2\pi(1-\Delta_1)n) = \cos(2\pi n - 2\pi\Delta_1 n) = \cos(2\pi\Delta_1 n)$$

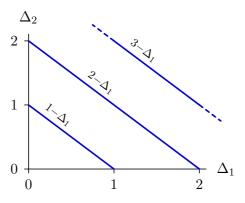
Points on this line represent pairs of sampling intervals (Δ_1 and Δ_2) that generate the same sequence of samples.

Similarly, the same sequence of samples results when $x(t) = \cos(2\pi t)$ is sampled at intervals Δ_1 or Δ_2 if $\Delta_2 = 2 - \Delta_1$.



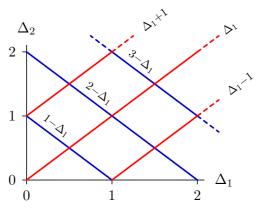
$$x[n] = \cos(2\pi\Delta_2 n) = \cos(2\pi(2-\Delta_1)n) = \cos(4\pi n - 2\pi\Delta_1 n) = \cos(2\pi\Delta_1 n)$$

Any integer shift also works.



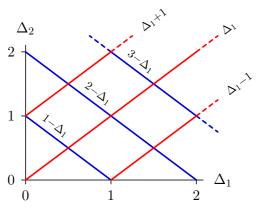
$$x[n] = \cos(2\pi\Delta_2 n) = \cos\left(2\pi(N - \Delta_1)n\right) = \cos(2N\pi n - 2\pi\Delta_1 n) = \cos(2\pi\Delta_1 n)$$

Sampling $x(t)=\cos(2\pi t)$ at $t=\Delta_1 n$ or $t=\Delta_2 n$ also generates the same sequence of samples when $\Delta_2=N+\Delta_1$.



$$x[n] = \cos(2\pi\Delta_2 n) = \cos\left(2\pi(N + \Delta_1)n\right) = \cos(2N\pi n + 2\pi\Delta_1 n) = \cos(2\pi\Delta_1 n)$$

Sampling $x(t) = \cos(2\pi t)$ at $t = \Delta_1 n$ or $t = \Delta_2 n$ also generates the same sequence of samples when $\Delta_2 = N + \Delta_1$.

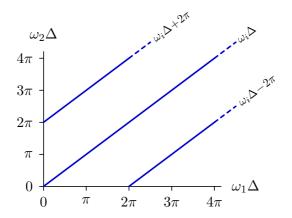


$$x[n] = \cos(2\pi\Delta_2 n) = \cos(2\pi(N + \Delta_1)n) = \cos(2N\pi n + 2\pi\Delta_1 n) = \cos(2\pi\Delta_1 n)$$

Many different sampling intervals result in the same sequence of samples. \rightarrow another special property of sinusoids

Sampling $\cos(\omega_1 t)$ and $\cos(\omega_2 t)$ with the same sampling interval Δ can also generate the same sequence of samples. For example, the same sequence of samples results if $\omega_2 \Delta = \omega_1 \Delta \pm 2\pi k$ for any integer value of k.

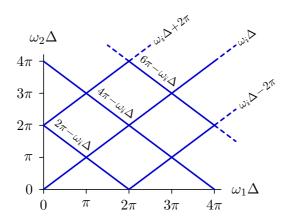
$$x[n] = \cos(\omega_2 \Delta n) = \cos((\omega_1 \Delta \pm 2\pi k)n) = \cos((\omega_1 \Delta)n)$$



Each point on the lines above show a pair of frequencies (ω_1 and ω_2) that generate the same sequence of samples: $x[n] = \cos(\omega_1 \Delta n) = \cos(\omega_2 \Delta n)$.

Sampling $\cos(\omega_1 t)$ and $\cos(\omega_2 t)$ with the same sampling interval Δ can also generate the same sequence of samples. As a second example, the same sequence of samples results if $\omega_2 \Delta = 2\pi k - \omega_1 \Delta$ for any integer value of k.

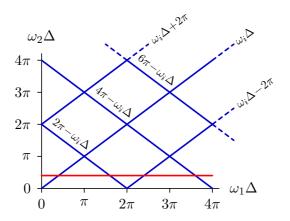
$$x[n] = \cos(\omega_2 \Delta n) = \cos((2\pi k - \omega_1 \Delta)n) = \cos((-\omega_1 \Delta)n) = \cos(\omega_1 \Delta n)$$



Each point on the lines above show a pair of frequencies (ω_1 and ω_2) that generate the same sequence of samples: $x[n] = \cos(\omega_1 \Delta n) = \cos(\omega_2 \Delta n)$.

Aliasing

Many input frequencies ω_1 generate the same output sequence of samples. For example, the same samples would result if the input frequency ω_1 times Δ were 0.4π or 1.6π or 2.4π or ... Therefore, it's impossible to determine what frequency produced an output at frequency 0.4π .

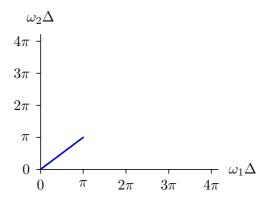


Since multiple frequencies ω_1 generate the same discrete samples, we say that these frequencies are **aliases** of each other.

Anti-Aliasing

We can prevent aliasing by removing **input** frequencies $\omega_1 \Delta > \pi$ and disregarding **output** frequencies $\omega_2 \Delta > \pi$.

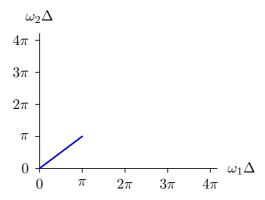
We call this low-frequency range of frequencies the baseband.



Anti-Aliasing

The maximum frequency that can be represented using this scheme is called the **Nyquist** frequency: $\omega_m = \pi/\Delta$, which equals half the sampling rate f_s .

$$f_m = \frac{\omega_m}{2\pi} = \frac{\pi/\Delta}{2\pi} = \frac{1}{2\Delta} = \frac{f_s}{2}$$



Consider 3 CT signals:

$$f_1(t) = \cos(4000t)$$
 ; $f_2(t) = \cos(5000t)$; $f_3(t) = \cos(6000t)$

Each of these is sampled so that

4. $f_3[n]$ $f_1[n]$ $f_2[n]$

$$f_1[n] = f_1(n\Delta)$$
 ; $f_2[n] = f_2(n\Delta)$; $f_3[n] = f_3(n\Delta)$

where $\Delta = 0.001$.

Which list goes from lowest to highest (baseband) frequency?

0.
$$f_1[n]$$
 $f_2[n]$ $f_3[n]$ 1. $f_1[n]$ $f_3[n]$ $f_2[n]$ 2. $f_2[n]$ $f_3[n]$ $f_3[n]$ 3. $f_2[n]$ $f_3[n]$ $f_1[n]$

5. $f_3[n]$ $f_2[n]$ $f_1[n]$

The CT signals are simple sinusoids:

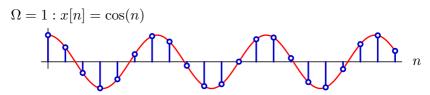
$$f_1(t) = \cos(4000t)$$
 ; $f_2(t) = \cos(5000t)$; $f_3(t) = \cos(6000t)$

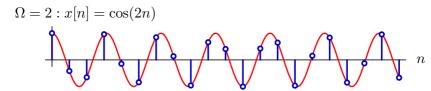
The DT signals are sampled versions ($\Delta = 0.001$):

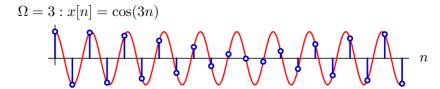
$$f_1[n] = \cos(4n)$$
 ; $f_2[n] = \cos(5n)$; $f_3[n] = \cos(6n)$

How do these discrete-time functions differ?

As frequency increases, the shapes of the sampled signals deviate from those of the underlying CT signals.



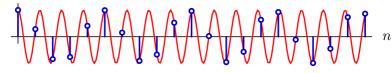




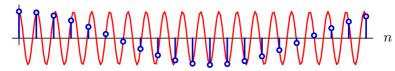
Worse and worse representation.

$$\Omega = 4: x[n] = \cos(4n) = \cos\left((2\pi - 4)n\right) \approx \cos(2.283n)$$

$$\Omega = 5 : x[n] = \cos(5n) = \cos((2\pi - 5)n) \approx \cos(1.283n)$$



$$\Omega = 6 : x[n] = \cos(6n) = \cos((2\pi - 6)n) \approx \cos(0.283n)$$



For $\Omega > \pi$, a lower frequency Ω_L has the same sample values as Ω .

$$\Omega = 4: x[n] = \cos(4n) = \cos\left((2\pi - 4)n\right) \approx \cos(2.283n)$$

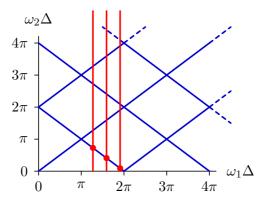
$$\Omega = 5: x[n] = \cos(5n) = \cos\left((2\pi - 5)n\right) \approx \cos(1.283n)$$

$$n$$

$$\Omega = 6: x[n] = \cos(6n) = \cos\left((2\pi - 6)n\right) \approx \cos(0.283n)$$

The same DT sequence represents multiple different values of Ω .

Graphically: As the input frequency $\omega_1\Delta$ goes from 4 to 5 to 6, the output baseband frequency decreases from approximately 2.3 to 1.3 to 0.3.



Consider 3 CT signals:

$$f_1(t) = \cos(4000t)$$
 ; $f_2(t) = \cos(5000t)$; $f_3(t) = \cos(6000t)$

Each of these is sampled so that

$$f_1[n] = f_1(n\Delta)$$
 ; $f_2[n] = f_2(n\Delta)$; $f_3[n] = f_3(n\Delta)$

where $\Delta = 0.001$.

- 0. $f_1[n]$ $f_2[n]$ $f_3[n]$ 1. $f_1[n]$ $f_3[n]$ $f_2[n]$
- 2. $f_2[n]$ $f_1[n]$ $f_3[n]$ 3. $f_2[n]$ $f_3[n]$ $f_1[n]$
- 4. $f_3[n]$ $f_1[n]$ $f_2[n]$ 5. $f_3[n]$ $f_2[n]$ $f_1[n]$

Anti-Aliasing Demonstration

Sampling Music.

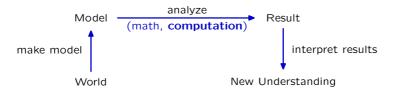
- ullet $f_s=11$ kHz without anti-aliasing
- \bullet $f_s=11$ kHz with anti-aliasing
- ullet $f_s=5.5$ kHz without anti-aliasing
- $f_s = 5.5$ kHz with anti-aliasing • $f_s = 2.8$ kHz without anti-aliasing
- $f_s = 2.8$ kHz with anti-aliasing

J.S. Bach, Sonata No. 1 in G minor Mvmt. IV. Presto Nathan Milstein, violin

Why does the aliased version (i.e., without anti-aliasing) sound so bad? Why is the anti-aliased version so much better?

Importance of Discrete Representations

Our goal is to develop **signal processing** tools to model interesting aspects of the world, to analyze the model, and to interpret the results.



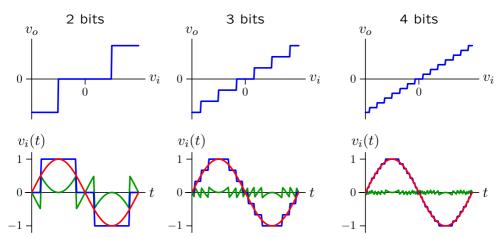
The increasing power and decreasing cost of computation makes the use of computation increasingly attractive.

However, many important signals are naturally described with continuous functions, that must be **sampled** in order to be analyzed computationally.

Today: understand relations between **continuous** and **sampled** signals.

Quantization

The information content of a signal depends not only with sample rate but also with the number of bits used to represent each sample.



Bit rate = $(\# bits/sample) \times (\# samples/sec)$

We hear sounds that range in amplitude from 1,000,000 to 1.

How many bits are needed to represent this range?

- 1. 5 bits
- 2. 10 bits
- 3. 20 bits
- 4. 30 bits
- 5. 40 bits

How many bits are needed to represent 1,000,000:1?

bits	range	
1	2	
$\frac{2}{2}$	$\begin{array}{c}2\\4\\8\\16\end{array}$	
3	8	
4	10	
9 6	$\begin{array}{c} 32 \\ 64 \end{array}$	
7	128	
8	256	
$egin{array}{c} 1 \\ 2 \\ 3 \\ 4 \\ 5 \\ 6 \\ 7 \\ 8 \\ 9 \\ \end{array}$	512	
10	1,024	
11	$\frac{2,048}{4,096}$	
$\frac{12}{12}$	4,096	
$\frac{13}{14}$	8, 192 16, 384	
$\begin{array}{c} 14 \\ 15 \end{array}$	$\frac{10,384}{32,768}$	
$\overset{15}{16}$	65,536	
17	131.072	
18	262,144	
19	$131,072 \\ 262,144 \\ 524,288 \\ 1,048,576$	
20	1,048,576	

We hear sounds that range in amplitude from 1,000,000 to 1.

How many bits are needed to represent this range? 3

- 1. 5 bits
- 2. 10 bits
- 3. 20 bits
- 4. 30 bits
- 5. 40 bits

Quantizing Music

- 16 bits/sample
- 6 bits/sample
- 5 bits/sample
- 4 bits/sample3 bits/sample
- 2 bit/sample

Quantizing Music

- 16 bits/sample
- 6 bits/sample
- 5 bits/sample
- 4 bits/sample3 bits/sample
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- 2 bit/sample

Quantizing Music

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Quantizing Music

- 16 bits/sample
- 6 bits/sample
- 5 bits/sample
- 4 bits/sample3 bits/sample
- 2 bit/sample

Converting an image from a continuous representation to a discrete representation involves the same sort of issues.

This image has 280×280 pixels, with brightness quantized to 8 bits.







8 bit image

7 bit image





8 bit image

6 bit image





8 bit image

5 bit image

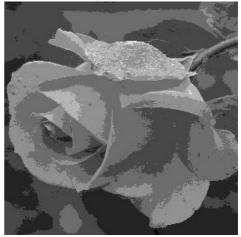




8 bit image

4 bit image

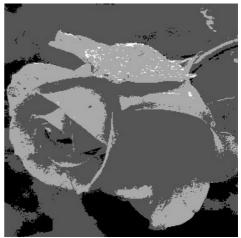




8 bit image

3 bit image





8 bit image

2 bit image





8 bit image

1 bit image

Quantizing Music With and Without (Robert's) Dither

- 4 bits/sample
- 4 bits/sample with dither
- 3 bits/sample
- 3 bits/sample with dither
- 2 bits/sample
- 2 bit/sample with dither

Quantizing Music With and Without (Robert's) Dither

- 4 bits/sample
- 4 bits/sample with dither
- 3 bits/sample
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- 2 bits/sample
- 2 bit/sample with dither

Quantizing Music With and Without (Robert's) Dither

- 4 bits/sample
- 4 bits/sample with dither
- 3 bits/sample
- 3 bits/sample with dither
- 2 bits/sample
- 2 bit/sample with dither

J.S. Bach, Sonata No. 1 in G minor Mvmt. IV. Presto Nathan Milstein, violin

In what way is the dithered version better?

Summary

We are highly motivated to develop discrete representations of signals – especially when they represent signals that are naturally described with continuous functions.

Information is generally lost in such discretization processes.

Today we discussed two mechanisms that can alter the information contained in a signal: **aliasing** and **quantization**.

Next time, we will develop representations that are specialized for discretetime signals.

Trig Table

```
sin(a+b) = sin(a) cos(b) + cos(a) sin(b)
sin(a-b) = sin(a) cos(b) - cos(a) sin(b)
cos(a+b) = cos(a) cos(b) - sin(a) sin(b)
cos(a-b) = cos(a) cos(b) + sin(a) sin(b)
tan(a+b) = (tan(a)+tan(b))/(1-tan(a) tan(b))
tan(a-b) = (tan(a)-tan(b))/(1+tan(a) tan(b))
sin(A) + sin(B) = 2 sin((A+B)/2) cos((A-B)/2)
sin(A) - sin(B) = 2 cos((A+B)/2) sin((A-B)/2)
cos(A) + cos(B) = 2 cos((A+B)/2) cos((A-B)/2)
cos(A) - cos(B) = -2 sin((A+B)/2) sin((A-B)/2)
sin(a+b) + sin(a-b) = 2 sin(a) cos(b)
sin(a+b) - sin(a-b) = 2 cos(a) sin(b)
cos(a+b) + cos(a-b) = 2 cos(a) cos(b)
cos(a+b) - cos(a-b) = -2 sin(a) sin(b)
2 \cos(A) \cos(B) = \cos(A-B) + \cos(A+B)
2 \sin(A) \sin(B) = \cos(A-B) - \cos(A+B)
2 \sin(A) \cos(B) = \sin(A+B) + \sin(A-B)
2 \cos(A) \sin(B) = \sin(A+B) - \sin(A-B)
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