Demonstration of Detection and Ranging Using Solvable Chaos

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ABSTRACT

Acoustic experiments demonstrate a novel approach to ranging and detection that exploits the properties of a solvable chaotic oscillator. This nonlinear oscillator includes an ordinary differential equation and a discrete switching condition. The chaotic waveform generated by this hybrid system is used as the transmitted waveform. The oscillator admits an exact analytic solution that can be written as the linear convolution of binary symbols and a single basis function. This linear representation enables coherent reception using a simple analog matched filter and without need for digital sampling or signal processing. An audio frequency implementation of the transmitter and receiver is described. Successful acoustic ranging measurements are presented to demonstrate the viability of the approach.

Keywords: chaos, matched filter, radar, acoustic ranging, solvable chaos

1. INTRODUCTION

The wide bandwidth and aperiodic properties of chaos naturally suggest benefits for high-resolution, unambiguous ranging in radar, sonar, and ladar systems [1-13]. An obvious, conventional approach might be to substitute chaos for the noise source in random-signal radar. In such a system, a segment of the transmitted waveform is sampled and stored, using a resolution defined by the signal bandwidth and the Nyquist sampling criterion. The stored signal is then used in a correlation receiver to detect a return signal and determine time of flight. The cross-correlations are usually done digitally, using a digital signal processor (DSP) and fast-Fourier transforms (FFT). In this approach, the distinguishing properties of a chaotic waveform are not used: chaos is simply a wide-bandwidth, random source.

In contrast, we recently developed an alternative approach to detection and ranging that truly exploits the properties of a chaotic waveform to alleviate the most expensive parts of random-signal radar—i.e., sampling, digital memory, and digital signal processor—while still maintaining the performance of a correlation receiver [13]. This new approach uses chaotic waveforms generated by an analytically solvable nonlinear oscillator comprising an ordinary differential equation and a discrete switching state [14-15]. This hybrid oscillator admits an exact solution, which can be written as the linear convolution of a symbolic dynamics and a basis function. This analytic representation is significant since it enables coherent reception using a simple analog matched filter and only a few stored symbols.

In this paper, we present recent acoustic experiments that demonstrate this approach to ranging and detection. For these experiments, an amplified speaker emits an audio-frequency waveform generated by an electronic realization of the hybrid oscillator. The transmitted waveform sounds like noise. A complementary receiver circuit incorporates a matched filter for the chaotic waveform, which is mathematically equivalent to a correlation receiver. At repeated intervals, a sequence of symbols detected in the symbolic dynamics of the emitted waveform is captured, thereby defining a transmitted signal for ranging. The captured symbol sequence is provided to the receiver, where it defines the matched filter for the transmitted signal. Practically, the symbols define weights applied to elements of a microphone array, the outputs of which are summed and passively filtered. The output of the matched filter is a continuous signal that is proportional to the cross-correlation of the transmitted and received signal. In operation, a consistent peak in the output of the matched filter is observed, which indicates a detected target. Ranging is derived by the time of flight for the consistent peak.

The entire experimental system is realized using simple analog and digital electronic circuit components. Importantly, the receiver does not require waveform sampling or digital signal processing for detection. Real-time measurements using only an oscilloscope provide visible evidence of detection and ranging with the system.

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2. HYBRID CHAOTIC OSCILLATOR

The central element of the ranging demonstration system is the audio frequency oscillator shown schematically in Fig. 1. This electronic oscillator is a physical realization of a chaotic system previously considered by Tsubone and Saito [14] and Corron et al. [15]. This oscillator is a hybrid system, containing both an analog harmonic oscillator and digital logic circuits. For the acoustic system, we implemented the oscillator in an electronic circuit operating at roughly 10 kHz. The circuit is constructed using discrete analog and digital components on a solderless breadboard.

The dynamics of the oscillator are equivalent to a dimensionless hybrid model including a continuous scalar state \( v(t) \) and a discrete state \( s(t) \). The continuous-time dynamics are described by the differential equation

\[
\frac{d^2v}{dt^2} - 2\beta \frac{dv}{dt} + \left( \omega^2 + \beta^2 \right) \cdot (v - v_s) = 0
\]  

(1)

where \( \omega = 2\pi \) and \( 0 < \beta \leq \ln 2 \). Transitions in the discrete state are defined by the guard condition

\[
\frac{dv}{dt}(t) = 0 \quad \Rightarrow \quad v_s(t) = \begin{cases} +1 & v \geq 0 \\ -1 & v < 0 \end{cases}
\]  

(2)

meaning \( v_s(t) \) is set to the sign of \( v(t) \) whenever its time derivative vanishes, and \( v_s(t) \) maintains this value until the next critical point. This oscillator admits an exact, analytic solution

\[
\begin{align*}

v(t) &= \sum_{m=-\infty}^{\infty} s_m \cdot P(t-m) \\
v_s(t) &= \sum_{m=-\infty}^{\infty} s_m \cdot R(t-m)
\end{align*}
\]  

(3)

where each \( s_m = \pm 1 \) and \( P(t) \) and \( R(t) \) are fixed basis functions. In the solution, each symbol \( s_m \) modulates the fixed basis functions \( P(t) \) and \( R(t) \) centered at time \( t = m \). Thus, it is correct to think of the symbol \( s_m \) as the information emitted by the oscillator at time \( t = m \), and that the oscillator emits one symbol with each unit of time. The fixed basis functions are

\[
P(t) = \begin{cases} 
1 - e^{-\beta} e^{\beta t} \left( \cos \omega t - \frac{\beta}{\omega} \sin \omega t \right), & t < 0 \\
1 - e^{\beta(t-1)} \left( \cos \omega t - \frac{\beta}{\omega} \sin \omega t \right), & 0 \leq t < 1 \\
0, & 1 \leq t
\end{cases}
\]  

(4)

and

\[
R(t) = \begin{cases} 
0, & t < 0 \\
1, & 0 \leq t < 1 \\
0, & 1 \leq t
\end{cases}
\]  

(5)

which are shown in Fig. 2 for \( \beta = \ln 2 \).
The existence of fixed basis functions can be exploited for efficiently sampling the transmitted signal and storing a reference waveform. Here we show a transmitter design that achieves this important functionality.

The complete transmitter is shown in Fig. 3. At the center of the transmitter is the chaotic oscillator. The transmitted signal is the continuous state of the free-running chaotic oscillator, which is amplified and emitted by a conventional speaker. The additional circuitry at the bottom of the transmitter schematic derives a clock signal from the regular return times of the oscillator. The clock signal drives a binary shift register, which uses the signal $s(t)$ for the data input. A divide-by-$N$ counter circuit provides a signal to alternately enable and disable the shift register. For our experimental system, we typically use the value $N = 1024$.

In operation, the free-running oscillator generates a chaotic waveform that is continuously emitted from the speaker. While the shift register is enabled, the symbolic logic state $s(t)$ is sampled for each return and stored in the shift register. At any time, a fixed number of the most recent values of the logic state $s(t)$ are stored, which correspond to a sequence of amplitudes $s_m$ generated by the free-running oscillator. Although the figure only shows an eight-bit shift register, for the acoustic system we used a twelve-bit register.

After shifting $N$ successive returns through the register, the shift register is disabled by the signal from the divide-by-$N$ circuit. When disabled, the contents of the shift register are locked, thereby storing symbols that identify the signal transmitted just prior to the disabling transition. These stored symbols effectively define a reference waveform to use for detection and ranging in a correlation receiver. Compared to the usual Nyquist sampling criteria, this symbolic representation provides an order of magnitude reduction in the sampling and storage requirements for the reference waveform.
A simulated waveform and shift register content is shown in Fig. 4. The top plot shows the oscillator waveforms $v(t)$ and $v_s(t)$. The continuous waveform $v(t)$ is the transmitted waveform emitted by the speaker. The middle plot shows the clock signal that is extracted from the oscillator waveforms and defines the symbol timing. The bottom waveform shows the trigger signal derived by the divide-by-$N$ operation from the clock signal. The shift register is disabled by the low-to-high transition of the trigger signal. The dots in the top plot show the most recent eight symbols captured and stored by the shift register when it is disabled by the trigger. These symbols define a reference waveform for the correlation receiver.
4. CORRELATION RECEIVER

The second advantage in using an exactly solvable chaotic oscillator is the availability of a simple matched filter for the basis function [16]. This filter enables the construction of a simple correlation receiver for a chaotic waveform segment defined by truncated symbol sequence [13]. For the acoustic system, we implement a receiver using discrete circuit components and without requirement for a DSP.

The matched filter for a solvable chaotic waveform is

$$\frac{dp}{dt} = \sum_{m=1-N}^{0} s_m \cdot \left\{ \tilde{v} (t+1+m) - \tilde{v} (t+m) \right\}$$

$$\frac{d^2 \xi}{dt^2} + 2\beta \frac{d\xi}{dt} + (\omega^2 + \beta^2) (\xi - \eta) = 0$$

where $\tilde{v}$ is the received signal, $\eta$ is an intermediate state, and $\xi$ is the output of the matched filter [13]. The finite symbol sequence $s_m$ for $m = 1-N, ..., 0$ defines the reference waveform for the filter. The function of the matched filter is mathematically equivalent to a correlation with the reference waveform. For a signal corrupted by additive white Gaussian noise (AWGN), the matched filter is the optimal linear receiver for detecting the signal [17].

Fig. 5 shows a schematic implementation of the matched filter for the acoustic ranging system. The received waveform impinges on a microphone array, shown at top. The spacing of the microphones in the array is chosen to realize the evenly spaced time delays in the first equation of the matched filter. Differential amplifiers between adjacent microphones provide the difference signal of successively lagged signals, which are multiplied by ±1 according to the symbols defining the reference waveform. The summed differences are integrated and drive the harmonic filter to generate the matched filter output.

5. RANGING SYSTEM

The transmitter and receiver were constructed and installed in an acoustically anechoic chamber for demonstration and test. The speaker was positioned at different ranges in front of the microphone array, as shown in Fig. 6. A handheld oscilloscope, triggered on the disable signal to the shift register, was used to monitor the receiver output. The transmitter and matched filter states were also connected to a computer for instrumentation and tuning purposes.
In operation, a consistent spike in the matched filter output was evident at a delay corresponding to the distance from speaker to microphone array. In a single return for one instance of the reference waveform, this peak may be indistinguishable from background noise, intersymbol interference, or waveform sidelobes. However, the consistent peak emerges when multiple receiver outputs are averaged, which was conveniently provided by the oscilloscope. Typical outputs for 64 averaged returns are shown in Fig. 7, which shows the detection and correctly estimated range at three speaker positions.

6. CONCLUSIONS
We successfully demonstrated acoustic ranging using a simple and inexpensive experimental system by exploiting the properties of exactly solvable chaos. Notably, correlation receiver capability is achieved using analog hardware, and significant pulse compression is obtained without the sampling and storage requirements of a comparable random signal waveform. This successful demonstration enables the development of new, low-cost sonar and radar technologies using chaotic waveforms.
REFERENCES


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Chaotic Waveforms Appear Well Suited for Radar Applications...
Chaos Noise Radar

\[ r = \frac{v \cdot \Delta t}{2} \]

- Chaotic Oscillator
- Analog-to-Digital Converter
- Digital Memory
- Analog-to-Digital Converter
- DSP-FFT Correlator
Designing a Waveform

- Random bit sequence

\[ \{\ldots, s_{-2}, s_{-1}, s_0, s_1, s_2, \ldots\} \quad s_m = \pm 1 \]

- Encoded waveform

\[ u(t) = \sum_{m=-\infty}^{\infty} s_m \cdot \rho(t - m) \]

- Basis function

\[ \rho(t) = ??? \]
Receiver Constraints

• Want correlation performance
• Cannot afford digital processing

• What if we use a very simple, analog matched filter?

\[ M_1 \ast \delta(t) = \rho(-t) \]

\[ \rho(t) = \begin{cases} 
-\frac{\omega^2 + \beta^2}{\omega} \cdot \sin(\omega t) \cdot 2^t, & t < 0 \\
0, & t \geq 0 
\end{cases} \]

Note: RLC values selected to meet “tuning” requirements: \( \beta = \ln 2 \), \( \omega = 2\pi \)
Transformed Basis Function

\[ \delta(t) \]

\[ \rho(t) \]

Integrate and Dump

\[ \phi(t) \]

\[ P(t) \]
Linear Synthesis:

\[ v(t) = \sum_{m=-\infty}^{\infty} s_m \cdot P(t - m) \]

\[ v_s(t) = \sum_{m=-\infty}^{\infty} s_m \cdot R(t - m) \]

\[ s_m = \pm 1 \]

\[ \beta = \ln 2 \]
Chaotic Oscillator

- Hybrid oscillator circuit (analog and digital components):

\[ s = \{T, F\} \]
\[ v_s(t) = \pm 1 \text{ V} \]
Transmitter

\[ \frac{dv}{dt} < 0 \]

\[ v > 0 \]

\[ v(t) \]

Oscillator
Transmitter

Transmitted Waveform

Oscillator

Speaker

\[ \frac{dv}{dt} < 0 \]

\[ v(t) \]

\[ v_s(t) \]

\[ s \]
Transmitter

Transmitted Waveform

Oscillator

Regular Clock Signal Derived From Chaotic Oscillator

Shift Register Stores Symbols in Transmitted Waveform

Trigger Defines Start of Range Measurement

Speaker

\[ v(t) = \sum_{m=-\infty}^{\infty} s_m \cdot P(t - m) \]

To Receiver
Shift Register

Frozen Contents of Shift Register when Disabled

Shift Register Remembers Transmitted Waveform at Trigger Event

\[ v_{ref}(t) = \sum_{m=1}^{0} s_m \cdot P(t-m) \]

From Transmitter

\[ s_0 = -1 \]
\[ s_{-1} = -1 \]
\[ s_{-2} = +1 \]
\[ s_{-3} = +1 \]
\[ s_{-4} = -1 \]
\[ s_{-5} = +1 \]
\[ s_{-6} = -1 \]
\[ s_{-7} = +1 \]
Linear Convolution Waveform

\[ v_{ref}(t) = \sum_{m=1-N}^{0} s_m \cdot P(t - m) \]

Symbols Used to Store Reference Waveform

Receiver Can Exploit Basis Function for Detection
Symbols Used to Store Reference Waveform

$$v_{ref} (t) = \sum_{m=1-N}^{0} s_m \cdot P(t - m)$$

Receiver Can Exploit Basis Function for Detection

$$v_{ref} (t) = \sum_{m=1-N}^{0} s_m \cdot P(t - m)$$
$v_{ref}(t) = P(t)$

**Matched Filter**

**Integrate and Dump:**

$$\frac{d\eta}{dt} = \tilde{v}(t+1) - \tilde{v}(t)$$

**Harmonic Filter:**

$$\frac{d^2 \xi}{dt^2} + 2\beta \frac{d\xi}{dt} + (\omega^2 + \beta^2)\xi = (\omega^2 + \beta^2) \cdot \eta(t)$$
Matched Filter

$$v_{ref}(t) = P(t)$$

**Matched Filter**

**Integrate and Dump:**

$$\frac{d \eta}{dt} = \bar{v}(t + 1) - \bar{v}(t)$$

**Harmonic Filter:**

$$\frac{d^2 \xi}{dt^2} + 2 \beta \frac{d \xi}{dt} + (\omega^2 + \beta^2) \xi = (\omega^2 + \beta^2) \cdot \eta(t)$$
Correlation Receiver

\[ v_{ref}(t) = \sum_{m=0}^{0} s_m \cdot P(t-m) \]

From Transmitter

\[ \frac{d\eta}{dt} = \sum_{m=0}^{0} s_m \cdot \{ \tilde{v}(t+1+m) - \tilde{v}(t+m) \} \]
Correlation Receiver

\[ v_{ref}(t) = \sum_{m=-1}^{0} s_m \cdot P(t - m) \]

From Transmitter

\[ \frac{d\eta}{dt} = \sum_{m=-1}^{0} s_m \cdot \{ \tilde{v}(t + 1 + m) - \tilde{v}(t + m) \} \]
Correlation Receiver

\[ v_{ref}(t) = \sum_{m=-2}^{0} s_m \cdot P(t - m) \]

From Transmitter

\[ \frac{d\eta}{dt} = \sum_{m=-2}^{0} s_m \cdot \{\tilde{v}(t + 1 + m) - \tilde{v}(t + m)\} \]
Correlation Receiver

\[ v_{\text{ref}}(t) = \sum_{m=-3}^{0} s_m \cdot P(t - m) \]

From Transmitter

\[ \frac{d\eta}{dt} = \sum_{m=-3}^{0} s_m \cdot \{ \tilde{v}(t + 1 + m) - \tilde{v}(t + m) \} \]
Correlation Receiver

\[ v_{ref}(t) = \sum_{m=-4}^{0} s_m \cdot P(t - m) \]

From Transmitter

\[ \frac{d\eta}{dt} = \sum_{m=-4}^{0} s_m \cdot \{ \tilde{v}(t + 1 + m) - \tilde{v}(t + m) \} \]
Correlation Receiver

\[
v_{ref}(t) = \sum_{m=-5}^{0} s_m \cdot P(t-m)
\]

Microphones

From Transmitter

\[
d\eta = \sum_{m=-5}^{0} s_m \cdot \{\tilde{\nu}(t+1+m) - \tilde{\nu}(t+m)\}
\]

Matched Filter

Integrate and Dump

Harmonic Filter

\[\xi(t)\]
Correlation Receiver

\[ v_{ref}(t) = \sum_{m=-6}^{0} s_m \cdot P(t - m) \]

Microphones

From Transmitter

\[ \frac{d\eta}{dt} = \sum_{m=-6}^{0} s_m \cdot \left\{ \tilde{v}(t + 1 + m) - \tilde{v}(t + m) \right\} \]
Correlation Receiver

\[ v_{\text{ref}}(t) = \sum_{m=-7}^{0} s_m \cdot P(t - m) \]

From Transmitter

\[ \frac{d\eta}{dt} = \sum_{m=-7}^{0} s_m \cdot \{ \tilde{v}(t + 1 + m) - \tilde{v}(t + m) \} \]
Ranging System

\[ r = \frac{v \cdot \Delta t}{2} \]
Simulation Waveform 1

\[ v(t), s(t), \xi(t) \]

\[ t \]

\[ \xi(t) \]

CLK

TRIG

Simulation waveforms showing various parameters over time.
Simulation Waveform 2
Simulation Waveform 3
Simulation Waveform 4

v(t), s(t)

CLK

TRIG

ξ(t)

-20 -10 0 10 20

-10 0 10 20

-2 0 2

-20 -10 0 10 20

-10 0 10

v(t), s(t), CLK, TRIG, ξ(t) vs t
Simulation Waveform 6

\[ v(t), s(t) \]

CLK

TRIG

\[ \xi(t) \]

\[ -20 \quad -10 \quad 0 \quad 10 \quad 20 \]

\[ -10 \quad 0 \quad 10 \quad 20 \]

\[ t \]
Simulation Waveform 7

\[ v(t), s(t) \]

\[ \xi(t) \]

CLK

TRIG

\[ t \]
Simulation Waveform 8

\[ v(t), s(t) \]

-2

CLK

TRIG

\[ \xi(t) \]

-10

0

10

20

-20

-10

0

10

20

\[ t \]
Simulation Waveform 9
Simulation Averaged Matched Filter
Experimental System

Microphone Array

~60 cm

Speaker
Experimental System

oscillator clock ~10 kHz → 3-cm microphone spacing
oscilloscope averaging
Measured Waveform 1

\[ v(t) \ (V) \]

\[ \xi(t) \ (V) \]

\[ t \ (s) \]
Measured Waveform 2

\[ v(t) \ (V) \]

\[ \xi(t) \ (V) \]

\( t \ (s) \)
Measured Waveform 3

\[ v(t) \ (V) \]

\[ \xi(t) \ (V) \]

\( t \ (s) \)
Measured Waveform 4

The top graph represents the measured voltage waveform $v(t)$ (V) as a function of time $t$ (s). The waveform shows a series of oscillations with varying amplitudes and frequencies.

The bottom graph represents the measured voltage waveform $\tilde{v}(t)$ (V) as a function of time $t$ (s). This waveform has a different scale and possibly a different reference level compared to the top graph, indicating a phase shift or different time scale.

The graphs are overlaid with annotations and markers to highlight specific features or data points of interest.
Measured Waveform 5

\[ v(t) \text{ (V)} \]

\[ \theta(t) \text{ (V)} \]

\[ t \text{ (s)} \]

\[ -0.001 \quad 0.000 \quad 0.001 \quad 0.002 \quad 0.003 \]

\[ -5 \quad -4 \quad -3 \quad -2 \quad -1 \quad 0 \quad 1 \quad 2 \quad 3 \quad 4 \quad 5 \]
Measured Waveform 6

v(t) (V)

\( v(t) \) vs. \( t \) (s)

\( \xi(t) \) (V)

\( \xi(t) \) vs. \( t \) (s)
Measured Waveform 7

\[ v(t) \ (V) \]

\[ \xi(t) \ (V) \]

\[ t \ (s) \]
Measured Waveform 8

\[ v(t) \ (V) \]

\[ \frac{\dot{g}(t)}{g(t)} \ (V) \]

\[ t \ (s) \]
Measured Waveform 9
Measured Average Matched Filter

\[ r = v \cdot \Delta t \]

\[ \Delta t = 0.00174 \text{ s} \]

\[ r = 59 \text{ cm} \]

\[ v = 340 \text{ m/s} \]
Range Measurements

Varied Speaker Positions:

- $r = 81$ cm
- $r = 59$ cm
- $r = 43$ cm
Noise Radar

Noise Source

Reference Waveform

Analog-to-Digital Converter

Digital Memory

Analog-to-Digital Converter

DSP-FFT Correlator

\[ r = \frac{v \cdot \Delta t}{2} \]
Chaos Noise Radar

\[ r = \frac{v \cdot \Delta t}{2} \]

\[ \Delta t \]

Chaos Oscillator

Reference Waveform

Analog-to-Digital Converter

Digital Memory

Analog-to-Digital Converter

DSP-FFT Correlator

\[ t \]
Chaos Noise Radar

\[ r = \frac{v \cdot \Delta t}{2} \]
Chaos Radar

\[ r = \frac{v \cdot \Delta t}{2} \]
Conclusions

• Acoustic Ranging Using Solvable Chaos
  – Hybrid Oscillator
    • binary symbols
    • basis pulse
  – Matched Filter
  – Exploits Chaos in Meaningful Way
  – Simple, Low-Cost Components

• Concept for Low-Cost UWB Radar
  – No Analog-to-Digital Conversion
  – No DSP-FFT Correlations
  – Reduced Sampling and Storage
  – Matched Filter Performance
• Square pulse

\[ \phi(t) = \begin{cases} 
1, & 0 \leq t < 1 \\
0, & \text{otherwise} 
\end{cases} \]

• Linear transformation

\[ P(t) = \int_{-\infty}^{\infty} \phi(\tau) \cdot \rho(t-\tau) d\tau \]

• Matched filter

\[ M_2 \circ \rho(-t) = P(-t) \]

\[ M_2 \circ (M_1 \circ \delta(t)) = P(-t) \]
Differential Equation

• Waveform...

\[ v(t) = \sum_{m=-\infty}^{\infty} s_m \cdot P(t-m) \]

... satisfies the differential equation

\[ \frac{d^2v}{dt^2} - 2\beta \frac{dv}{dt} + \left( \omega^2 + \beta^2 \right)(v - v_s) = 0 \]

... where

\[ v_s(t) = \sum_{m=-\infty}^{\infty} s_m \cdot \phi(t-m) \]
Guard Condition

- Waveform also satisfies...

\[ \frac{dv}{dt} = 0 \implies v_s = \text{sgn}(v) \]
Chaos

• Interpret as a dynamical system...

\[
\frac{d^2 v}{dt^2} - 2\beta \frac{dv}{dt} + \left(\omega^2 + \beta^2\right) \cdot (v - v_s) = 0
\]

\[
\frac{dv}{dt} = 0 \quad \Rightarrow \quad v_s = \text{sgn}(v)
\]

Oscillator Model

\[
\frac{d^2 v}{dt^2} - 2\beta \frac{dv}{dt} + \left(\omega^2 + \beta^2\right)(v - v_s) = 0
\]

\[
\frac{dv}{dt} = 0 \quad \Rightarrow \quad v_s = \text{sgn}(v)
\]

Analytic Solution

• Model admits an exact solution (easily verified):

\[ v(t) = \sum_{m=-\infty}^{\infty} s_m \cdot P(t - m), \quad v_s(t) = \sum_{m=-\infty}^{\infty} s_m \cdot R(t - m) \]

where each \( s_m = \pm 1 \) and...

\[ P(t) = \begin{cases} \left(1 - e^{-\beta}\right) e^{\beta t} \left(\cos \omega t - \frac{\beta}{\omega} \sin \omega t\right), & t < 0 \\ 1 - e^{\beta(t-1)} \left(\cos \omega t - \frac{\beta}{\omega} \sin \omega t\right), & 0 \leq t < 1 \\ 0, & 1 \leq t \end{cases} \]

\[ R(t) = \begin{cases} 0, & t < 0 \\ 1, & 0 \leq t < 1 \\ 0, & 1 \leq t \end{cases} \]
Symbols vs. Nyquist Sampling

- **Nyquist-Shannon sampling requirement**
  
  \[ f_s > 2B \]
  
  \[ N_b = 8, 12, 16, ... \ ??? \]

- **Symbolic representation**
  
  \[ v(t) = \sum_{m=-\infty}^{\infty} s_m \cdot P(t-m) \]

  \[ f_s = B \]
  
  \[ N_b = 1 \]
Correlation Receiver

- Channel Model

\[ v(t - \Delta t) \rightarrow \tilde{v}(t) \rightarrow \text{Receiver} \rightarrow \xi(t) \]

- Correlation Receiver
  - optimal linear filter for detecting a given signal in additive white Gaussian noise (AWGN)

\[ \xi(t) = \int_{-\infty}^{\infty} \tilde{v}(\tau) \cdot v_{\text{ref}}(\tau - t) \, d\tau \]

\[ v_{\text{ref}}(t) = \sum_{m=1-N}^{0} s_m \cdot P(t - m) \]
• Correlation Receiver

\[
\xi(t) = \int_{-\infty}^{\infty} \tilde{v}(\tau) \cdot v_{ref}(\tau - t) \, d\tau
\]

\[
= \int_{-\infty}^{\infty} \tilde{v}(\tau) \cdot \left\{ \sum_{m=1-N}^{0} s_m \cdot P(\tau - t - m) \right\} \, d\tau
\]

\[
= \sum_{m=1-N}^{0} s_m \cdot \int_{-\infty}^{\infty} \tilde{v}(\tau) \cdot P(\tau - t - m) \, d\tau
\]

\[
= \sum_{m=1-N}^{0} s_m \cdot \int_{-\infty}^{\infty} \tilde{v}(\tau + m) \cdot P(\tau - t) \, d\tau
\]

\[
\xi(t) = \int_{-\infty}^{\infty} \left\{ \sum_{m=1-N}^{0} s_m \cdot \tilde{v}(\tau + m) \right\} \cdot P(\tau - t) \, d\tau
\]
Matched Filter for Basis Function

- Linear Filter

\[ L \circ \tilde{v}(t) = \int \tilde{v}(\tau) P(\tau - t) \, d\tau \]

- impulse response is time reversal of basis function...

- Basis Function
  \[ P(t) \]

- Matched Filter
  \[ L \circ \delta(t) = P(-t) \]
Consider the driven, damped harmonic, linear filter...

\[ \frac{d^2 \xi}{dt^2} + 2\beta \frac{d\xi}{dt} + \left( \omega^2 + \beta^2 \right) \xi = \left( \omega^2 + \beta^2 \right) \cdot \begin{cases} 1, & -1 \leq t < 0 \\ 0, & \text{otherwise} \end{cases} \]

\[ \xi(t) = P(-t) \]

**Matched Filter:**

\[ \frac{d\eta}{dt} = \tilde{v}(t+1) - \tilde{v}(t) \]

\[ \frac{d^2 \xi}{dt^2} + 2\beta \frac{d\xi}{dt} + \left( \omega^2 + \beta^2 \right) \xi = \left( \omega^2 + \beta^2 \right) \cdot \eta(t) \]

**Integrate and Dump**

Transforms impulse to square pulse

**Harmonic Filter**

Second-order damped oscillator
**Matched Filter:**

\[
\frac{d\eta}{dt} = \tilde{v}(t+1) - \tilde{v}(t)
\]

\[
\frac{d^2\xi}{dt^2} + 2\beta \frac{d\xi}{dt} + \left(\omega^2 + \beta^2\right) \xi = \left(\omega^2 + \beta^2\right) \cdot \eta(t)
\]

- **Impulse Response**

\[
\tilde{v}(t) = \delta(t) \quad \Rightarrow \quad \eta(t) = \phi(-t) \quad \xi(t) = P(-t)
\]
Correlation Receiver

\[ v_{\text{ref}}(t) = P(t) \]

\[ v_{\text{ref}}(t) = \sum_{m=1-N}^{0} s_m \cdot P(t - m) \]

**Matched Filter**

Integrate and Dump:

\[ \frac{d\eta}{dt} = \bar{v}(t + 1) - \bar{v}(t) \]

Harmonic Filter:

\[ \frac{d^2 \xi}{dt^2} + 2\beta \frac{d\xi}{dt} + \left( \omega^2 + \beta^2 \right) \xi = \left( \omega^2 + \beta^2 \right) \eta(t) \]

**Correlation Receiver**

Integrate and Dump:

\[ \frac{d\eta}{dt} = \sum_{m=1-N}^{0} s_m \cdot \{ \bar{v}(t + 1 + m) - \bar{v}(t + m) \} \]

Harmonic Filter:

\[ \frac{d^2 \xi}{dt^2} + 2\beta \frac{d\xi}{dt} + \left( \omega^2 + \beta^2 \right) \xi = \left( \omega^2 + \beta^2 \right) \eta(t) \]

Reference Waveform
Correlation Receiver

\[ \frac{d\eta}{dt} = \sum_{m=1-N}^{0} s_m \cdot \{ \tilde{v}(t + 1 + m) - \tilde{v}(t + m) \} \]