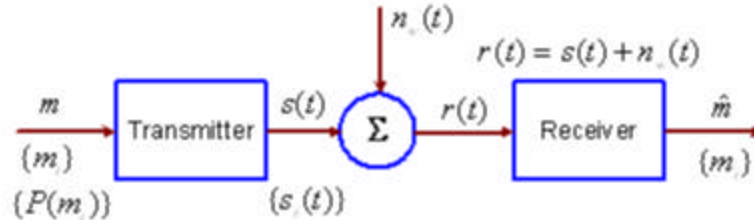


## 2. OPTIMUM RECEIVER PRINCIPLES

### 2.1 Maximum *Aposteriori* Receiver

Consider the generic block diagram of end-to-end communication over the ubiquitous additive white Gaussian noise (AWGN) channel.



- Source:  $\{m_i\}$  with *a priori* probabilities:  $\{P(m_i)\}$
- Transmitter: A particular message symbol is represented by a signal waveform allowable in the signal space permitted for a given modulation technique.

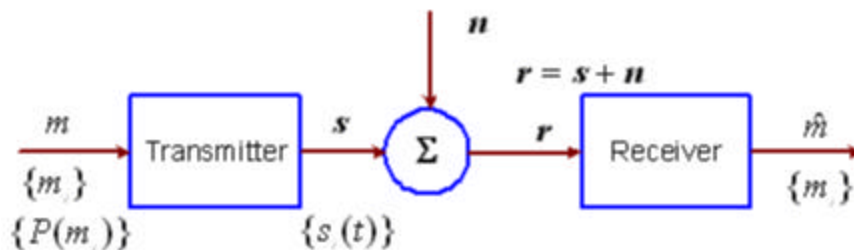
$$m = m_i \Leftrightarrow s(t) = s_i(t) \quad (2.1)$$

- Channel:  $r(t) = s(t) + n_w(t)$  (2.2)

**Problem 1:** Design an optimum receiver which estimates  $\hat{m}$  for the transmitted signal  $s(t)$  of a source output  $m$  such that the probability of error  $P(e) \equiv \Pr ob(\hat{m} \neq m)$  is **MINIMUM**.

**Problem 2:** Given that  $\{P(m_i)\}$  are UNKNOWN, which is the real-life problem in many emerging communication systems, design a similar optimum receiver. (Inherently more difficult task!).

**VECTOR CHANNEL:** Consider the case when a sequence of source outputs are bundled into a vector form and transmitted as the case of QAM and other many signaling schemes. In some cases, the signal itself may be in a vector form to start with, as in the case of LPC coefficients in a CELP Speech coder.



- Source Information is mapped into source vectors:  $\{\underline{s}_i; i = 0, 1, \dots, M-1\}$ , where each vector is composed of N-components:  $\underline{s}_i = [s_{i1}, s_{i2}, \dots, s_{iN}]$ .
- Received Information is also mapped into vectors:

$$\underline{r}_i = \underline{s}_i + \underline{n}_i = [s_{i1} + n_{i1}, s_{i2} + n_{i2}, \dots, s_{iN} + n_{iN}] \quad (2.3)$$

Given that the received vector is a point  $\underline{r} = \underline{r}$  in the N-dimensional space with coordinates:

$\underline{s} = [s_1, s_2, \dots, s_N]$  then the optimum receiver must compute the transmitted vector signal  $\underline{s}_i$  for the message  $m_i$  having a maximum a posteriori probability from its knowledge of the set of parameters:  $P_{\underline{s}}$ ,  $\{\underline{s}_i\}$ , and the source distribution:  $\{Pr ob(m_i)\}$ .

In other words:

$$\hat{m} = m_k \quad \text{if} \quad Pr ob(m_k | \underline{r} = \underline{r}) > Pr ob(m_i | \underline{r} = \underline{r}) \quad \text{for all } i \neq k \quad (2.4)$$

which is a nearly impossible challenge to meet in many real-life situations.

### Do we have an equivalent task?

Consider the correct decision for a given incoming vector:

$$Pr ob(C | \underline{r} = \underline{r}) = Pr ob(m_k | \underline{r} = \underline{r}) \quad (2.5)$$

and the overall correct decision is simply ensemble of correct decisions:

$$Pr ob(C) = \int_{-\infty}^{\infty} Pr ob(C | \underline{r} = \underline{r}) \cdot P_{\underline{r}}(\underline{r}) d\underline{r} \quad (2.6)$$

Since  $P_{\underline{r}}(\underline{r}) \geq 0$  we do not need to include it in the maximization process, i.e. only the term

$Pr ob(C | \underline{r} = \underline{r})$  must be maximized. Let us use the Bayes Rule on (2.5)

$$Pr ob(m_i | \underline{r} = \underline{r}) = P(m_i) \cdot Pr ob_{\underline{r}}(\underline{r} | m_i) / P_{\underline{r}}(\underline{r}) \quad (2.7)$$

but the statement  $m = m_i$  is equivalent to  $\underline{s} = \underline{s}_i$  which implies:

$$Pr ob_{\underline{r}}(\underline{r} | m_i) = Pr ob_{\underline{r}}(\underline{r} | \underline{s} = \underline{s}_i) \quad (2.8)$$

Furthermore, the denominator term is independent of the index  $i$ , hence, the maximization and we have the revised principle for our optimum receiver:

$$\hat{m} = m_k \quad \text{if} \quad P(m_i) \cdot Pr ob_{\underline{r}}(\underline{r} | \underline{s} = \underline{s}_i) \quad \text{is maximum when } i = k \quad (2.9)$$

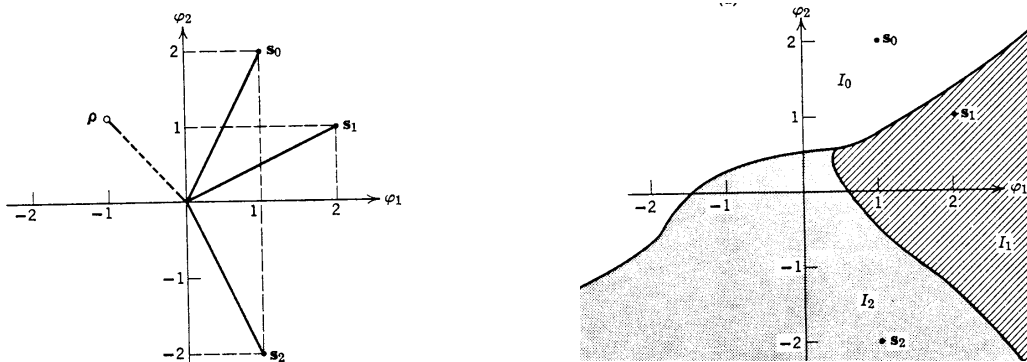
When  $P(m_i)$  are not known and the receiver can only maximize the last portion of (2.9). Then we have a restricted version of the general optimum receiver called **MAXIMUM-LIKELIHOOD (ML) Receiver**.

### ML Receiver Principle:

$$\hat{m} \Rightarrow m_k \quad \text{when} \quad Pr ob_{\underline{r}}(\underline{r} | \underline{s} = \underline{s}_i) \quad \text{is MAXIMUM.} \quad (2.10)$$

Decision Regions are needed to perform the mapping properly for each signal vector.

**Example 2.1:** Given (3) input vectors in a 2-D vector space with the following signal set assignment:  $m_0 \Rightarrow \underline{s}_0 = [1, 2]$ ;  $m_1 \Rightarrow \underline{s}_1 = [2, 1]$ ; and  $m_2 \Rightarrow \underline{s}_2 = [1, -2]$



Let us also assume that the input message probabilities:  $P(m_0), P(m_1), P(m_2)$  are given. For this assignment, our receiver will compute:

$$P(m_i).Pr_{ob_r}(\underline{r} | \underline{s} = \underline{s}_i) \quad \text{for } i=0,1,2.$$

**An ML Receiver will choose the index of the message with the largest product above.**

For every point  $\underline{r}$  in  $(\mathbf{j}_1, \mathbf{j}_2)$  plane an assignment can be made if the plane is partitioned into disjoint regions  $\{I_i\}$  for  $i=0,1,2$ ; which are called decision regions, very similar to the codeword selection process in Vector Quantization (VQ). Then we have the ML receiver as a simple geometric map:

$$\underline{r} = I_k \Rightarrow \hat{m} = m_k \text{ and an error is made if } \hat{m} \Rightarrow m_k \text{ iff } \underline{r} \notin I_k \quad (2.11)$$

## 2.2 ML Receiver for AWGN Channel

Given that the signal in the channel is corrupted by a zero mean AWGN with a variance  $\mathbf{s}^2$ .

$$\underline{r} = \underline{s} + \underline{n} = [s_1 + n_1, s_2 + n_2, \dots, s_N + n_N] \quad (2.12)$$

Now:

$$\underline{r} = \underline{r} \text{ when } \underline{s} = \underline{s}_i \text{ iff } \underline{n} = \underline{r} - \underline{s}_i \quad (2.13)$$

And then

$$P_r(\underline{r} | \underline{s} = \underline{s}_i) = P_n(\underline{r} - \underline{s}_i | \underline{s} = \underline{s}_i) \quad \text{for } i=0,1,\dots,M-1 \quad (2.14)$$

Since the signal  $\underline{s}$  and the channel noise  $\underline{n}$  are statistically independent  $P_{\underline{n}|\underline{s}} = P_{\underline{n}}$ . This simplifies (2.14) into:

$$P_n(\underline{r} - \underline{s}_i | \underline{s} = \underline{s}_i) = P_n(\underline{r} - \underline{s}_i) \quad (2.15)$$

In this case, the general ML decision function becomes  $P(m_i).P_n(\underline{r} - \underline{s}_i)$ . Now the components of signal is assumed to be independent, noise has a zero-mean we can write the noise distribution:

$$P_n(\underline{u}) = \frac{1}{(2p\mathbf{s}^2)^{N/2}} \exp\left\{-\frac{1}{2\mathbf{s}^2} \sum_{j=1}^N u_j^2\right\} \quad (2.16)$$

Let us use the following dot-product notation:

$$|\underline{u}|^2 = \underline{u} \bullet \underline{u}^* = \sum_{j=1}^N u_j^2$$

Our distribution is written as:

$$P_n(\underline{u}) = \frac{1}{(2p\mathbf{s}^2)^{N/2}} \exp\left\{-\frac{1}{2\mathbf{s}^2} |\underline{u}|^2\right\} \quad (2.17)$$

Then for this probability system we have the ML principle as:

$$\hat{m} \Rightarrow m_i \text{ whenever } P(m_i). \exp\left\{-\frac{1}{2\mathbf{s}^2} |\underline{r} - \underline{s}_i|^2\right\} \text{ is maximum.} \quad (2.18)$$

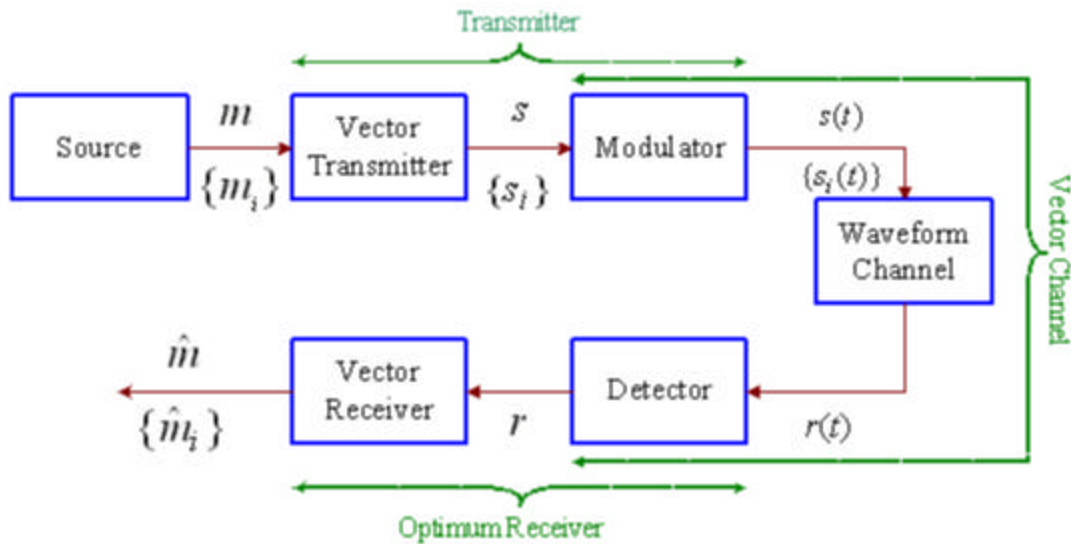
Equivalently, the task is to **MINIMIZE**:

$$|\underline{r} - \underline{s}_i|^2 - (2\mathbf{s}^2). \log_e P(m_i) \quad (2.19)$$

The first term is the Euclidean Distance between the received vector and a candidate signal vector. If all the messages are equally likely then the optimum decision rule does not depend on the index at all and we have the **MINIMUM MEAN-SQUARE (MMS) DISTANCE Receiver**. That is we assign the message index of the closest neighbor of the incoming vector, which is also known as the Nearest Neighbor Rule in VQ and other clustering techniques.

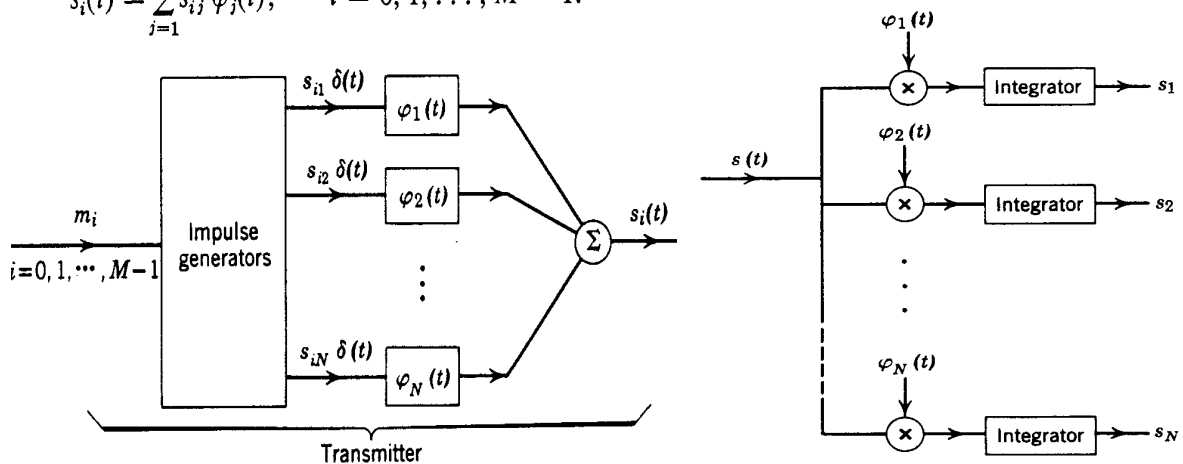
### 2.3 Correlation and Matched-Filter Receivers

If we revisit the Communication System Block Diagram for vector signals as shown below, it would be necessary to synthesize waveform signals to be transmitted over real-life channels, such as the twisted-pair or coaxial cable, microwave or fiber-optic links.



- It is necessary to synthesize the signal set  $\{s_i(t)\}$  at the transmitter. This can be achieved by "building blocks waveforms".
- Synthesis signal sets and Recovery of signal vectors:
  1. A set of N integrating filters are used to generate N signal components with strengths  $\{s_{ij}\}$ .
  2. The filter outputs are summed to yield the signal waveform:  $s(t)$  to be transmitted for a particular message  $m_i$  for each of M different messages.

$$s_i(t) = \sum_{j=1}^N s_{ij} \varphi_j(t); \quad i = 0, 1, \dots, M - 1.$$



$$s_i(t) = \sum_{j=1}^N s_{ij} \mathbf{j}_j(t) \quad \text{for } i=0,1,\dots,M-1 \quad (2.20)$$

1. Let us choose the building-block waveforms from an ortho-normal set such that:

$$\int_{-\infty}^{\infty} \mathbf{j}_j(t) \mathbf{j}_l(t) dt = \begin{cases} 1 & \text{if } j=l \\ 0 & \text{if } j \neq l \end{cases} \quad \text{for all } 1 \leq i, j \leq N \quad (2.21)$$

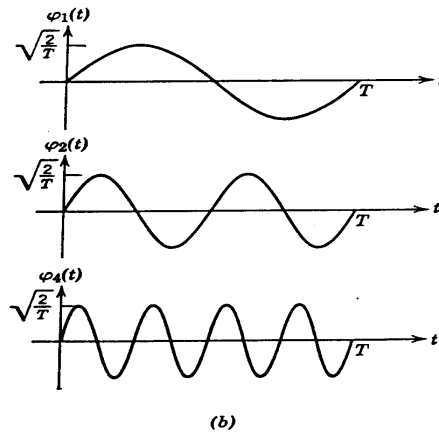
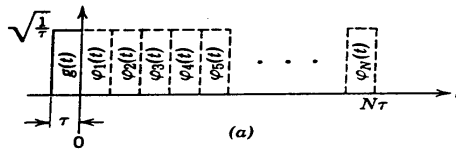
2. This will yield a probability of error independent of the actual wave-shapes.  
 3. We can exactly recover the signal vectors and hence, the messages in the absence of channel if we push these synthesized waveforms of (2.20) into a simple integrating filter structure as shown above.

$$\int s_i(t) \mathbf{j}_l(t) dt = \int \left[ \sum_{j=1}^N s_{ij} \mathbf{j}_j(t) \right] \mathbf{j}_l(t) dt = \sum_{j=1}^N s_{ij} \mathbf{d}_{jl} = s_{il} \quad (2.22)$$

If we perform similar integration for all the branches we obtain:  $\underline{s}_i = [s_{i1}, s_{i2}, \dots, s_{iN}]$ .

4. Examples of Orthonormal Signal Sets:

- Orthonormal time-shifted pulses:  $\mathbf{j}_j(t) = g(t - j\tau)$  for  $j=1,2,\dots,N$
- Orthonormal Fourier Transform pulses:  $\mathbf{j}_j(t) = \begin{cases} \sqrt{1/T} & -T \leq t < 0 \\ 0 & \text{otherwise} \end{cases}$



The optimum ML receiver of the system performs:

$$\text{Set } \hat{m} = m_k \quad \text{if } |\underline{r} - \underline{s}_i|^2 - 2\mathbf{S}_n^2 \log P(m_i) \text{ is MINIMUM.} \quad (2.23)$$

Square operations can be eliminated in (2.23) by observing:

$$|\underline{r} - \underline{s}_i|^2 = |\underline{r}|^2 - 2(\underline{r} \bullet \underline{s}_i) + |\underline{s}_i|^2 \quad (2.24)$$

where the dot product is given also by:

$$\underline{r} \bullet \underline{s}_i \equiv \sum_{j=1}^N r_j s_{ij} \tag{2.25}$$

**Observations :**

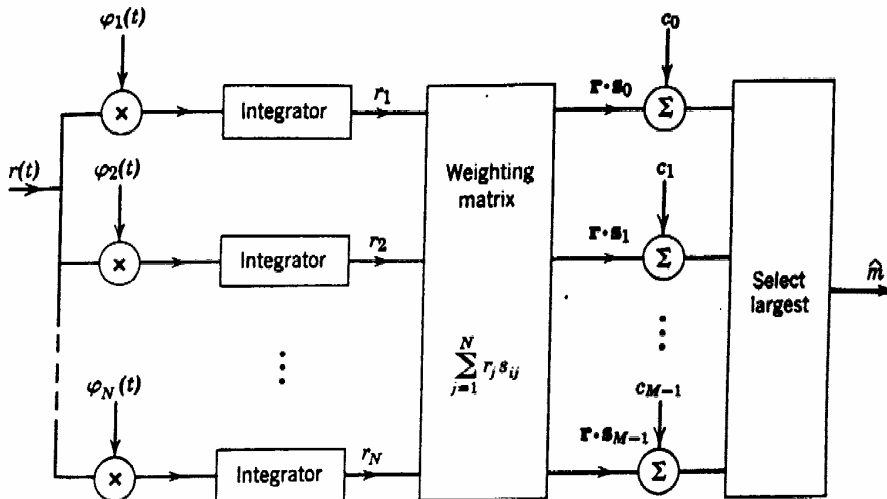
- Note 1: First term in (2.24) is independent of the index and no need to worry in optimization.
- Note 2: Last terms in (2.23) and (2.24) depend only on source side information supplied by the designer then they can be combined into a constant parameter set and burned into the ROM of the system:

$$c_i \equiv (1/2)[\mathbf{s}_n^2 \log P(m_i) - |\underline{s}_i|^2] \tag{2.26}$$

- The optimum receiver of (2.23) is now equivalent to:

$$\text{Set } \hat{m} = m_k \text{ if } (\underline{r} \bullet \underline{s}_i + c_i) \text{ is MAX.} \tag{2.27}$$

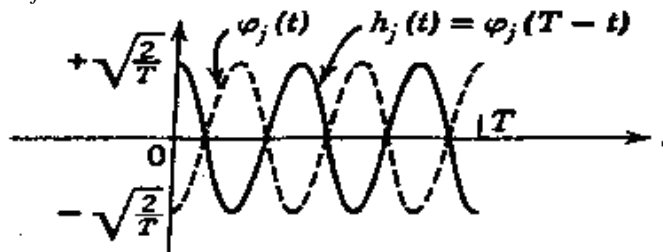
which is simply the structure of a **CORRELATION RECEIVER**.



**Note:** When the source vocabulary size M is not very large this implementation is not costly and most of the operations to the right of the "Integrators" can be done by table look-ups. However, when M is very large then the dot-products are usually handled by using "DSP" based devices. The use of multipliers can be avoided if we replace the structure to the left of the "Weighting Matrix" as follows:

1. Let us consider a filter with an impulse response

$$h_j(t) = j_j(T - t). \tag{2.28}$$



2. If the input to this filter is  $\underline{r}(t)$  then its response is simply:

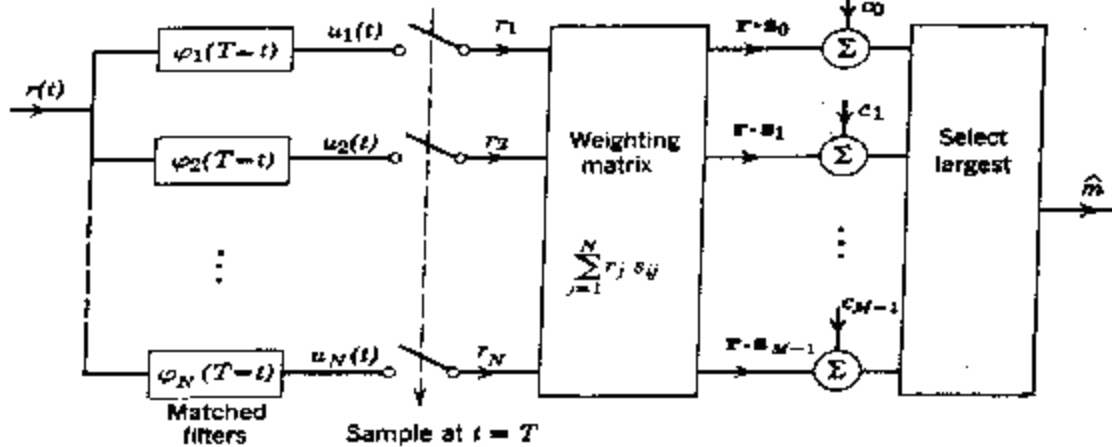
$$u_j(t) = \int_{-\infty}^{\infty} \underline{r}(\mathbf{a}) h(t - \mathbf{a}) d\mathbf{a} = \int_{-\infty}^{\infty} \underline{r}(\mathbf{a}) \mathbf{j}_j(T - t + \mathbf{a}) d\mathbf{a} \quad (2.29)$$

3. When we sample the output at  $t=T$  we have

$$u_j(T) \equiv r_j \quad (2.30)$$

4. Finally, the task is to push it through the weighting matrix and the rest of the receiver above.

This receiver is called a **"MATCHED-FILTER" Receiver** since it is constructed by using the shifted versions of the signal building block functions:  $\mathbf{j}_j(T - t)$ .



**Example 2.2:** Consider the case for a gated-sinusoidal tone signal with a gate period of  $T$  seconds as shown below. The convolution operation in the matched-filter above will result in a triangular enveloped sinusoid with the same frequency and thus it will peak at the sampling instant  $T$ .

