Pulse-Amplitude Modulation

ELEN 3024 - Communication Fundamentals

School of Electrical and Information Engineering, University of the Witwatersrand

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Pulse-Amplitude Modulation

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Barry et al., "Digital Communication", Chapter 5

5. Introduction

bit-streams inherently discrete-time, all physical media are continuous-time in nature

modulation \rightarrow bit stream represented as a continuous-time signalling

Consider PAM:

- baseband PAM
- passband transmission

5. Introduction (Continued)



Figure: Baseband



Figure: passband

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5. Introduction (Continued)

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Examples of PAM:

- PSK
- AM-PM
- QAM

5.1. Baseband PAM

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Baseband PAM transmitter sends information by modulating the amplitudes of a series of pulses:

$$s(t) = \sum_{m=-\infty}^{\infty} a_m g(t - mT)$$
 (1)

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ightarrow$ symbol rate

g(t)
ightarrow pulse shape

Set of amplitudes $\{a_m\} \rightarrow$ symbols

Signal \rightarrow sequence of possibly overlapping pulses \rightarrow amplitude of *m*'th pulse determined by *m*'th symbol

Equation (1) \rightarrow PAM, regardless of shape of g(t)

5.1. Baseband PAM

Example 5.1 Concepts:

1. Mapper \rightarrow converts input bit stream to modulating symbol stream

- a. In practice, symbols restricted to finite alphabet $\ensuremath{\mathcal{A}}$
- b. Convenient when $|\mathcal{A}| = 2^b$
- 2. transmit filter with impulse response g(t)

5.1. Baseband PAM

Note difference between baud rate (symbol rate) and bit rate

Assumption \rightarrow symbols from mapper independent and identically distributed, white discrete random process

Receiver \rightarrow recover transmitted symbols from a continuous-time PAM signal distorted by noisy channel

Assume for now noiseless PAM, in order to explore relationship between **bandwidth and symbol rate**

To recover the symbols $\{a_m\}$ from $s(t) \rightarrow \text{sample } s(t)$ at multiples of the symbol period

k-th sample:

$$s(kT) = \sum_{m=-\infty}^{\infty} a_m g(kT - mT) = a_m * g(kT)$$

Interpretation \rightarrow discrete-time convolution of the symbol sequence with a sampled version of the pulse shape

Decomposing the convolution sum into two parts:

$$s(kT) = g(0)a_k + \sum_{m \neq k} a_m g(kT - mT)$$

- First term \rightarrow desired signal
- Second term \rightarrow intersymbol interference (ISI)

$\mathsf{ISI} \to \mathsf{interference}$ from neighboring symbols

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When is no ISI present **OR** $s(kT) = a_k$?

When is no ISI present **OR** $s(kT) = a_k$?

When second term $\sum_{m \neq k} a_m g(kT - mT) = 0$

Alternatively:

 $g(kT) = \delta_k$

$$g(kT) = \delta_k \tag{2}$$

Taking Fourier transform on both sides and making use of sampling theorem:

$$\frac{1}{T}\sum_{m=-\infty}^{\infty}G\left(f-\frac{m}{T}\right)=1$$
(3)

Equation $3 \rightarrow Nyquist criterion$

Nyquist pulse \rightarrow satisfies Eq 3 (and Eq 2)

Nyquist criterion is the key that ties symbol rate to bandwidth

Nyquist criterion implies existence of a minimum bandwidth for transmitting at a certain symbol rate with no ISI

Alternatively, given certain bandwidth, maximum symbol rate for avoiding ISI.

Example 5-4

Sketch

Plot depicts $\frac{1}{T} \sum_{m=-\infty}^{\infty} G\left(f - \frac{m}{T}\right)$ for a particular pulse shape g(t) whose bandwidth is less than 1/(2T)

Effect of sampling \rightarrow place an image of G(f) at each multiple of the sampling rate.

Regardless of shape of $G(f) \rightarrow$ always gap between images whenever the pulse shape bandwidth is less than half the symbol rate

Such gaps prevent the images from adding to a constant

From example 5-4 evident minimum bandwidth required to avoid ISI is half the symbol rate 1/(2T)

Bandwidth of 1/(2T) eliminates gap between aliases

in order to ensure aliases add to a constant, each alias must itself have a rectangular shape, giving G(F):



Taking inverse Fourier transform \rightarrow minimum-bandwidth pulse satisfying Nyquist criterion:

$$g(t) = \frac{\sin(\pi t/T)}{\pi t/T}$$

Refer to sketch

Observe that pulse has zero crossings at all multiples of T except at t = 0, where g(0) = 1

Example 5-5.

Using sinc pulses, transmit $a_0 = 1$ and $a_1 = 2$

Sketch resulting signal

Nyquist criterion implies a maximum symbol rate for a given bandwidth

If we are constrained to frequencies |f| < W, the maximum symbol rate that can be achieved with zero ISI is 1/T = 2W

Minimum bandwidth is desirable, but the ideal bandlimited pulse is impractical

The bandwidth W of a practical pulse is larger than its minimum value by a factor $1 + \alpha$:

$$W = \frac{1+\alpha}{2T}$$

 $\alpha \rightarrow {\rm excess-bandwidth}$ parameter

Excess bandwidth also expressed as percentage, 100 % $\rightarrow \alpha = 1$ and bandwidth of 1/T (twice the minimum bandwidth)

Practical systems, excess bandwidth in range of 10 % to 100 %

Increasing the excess bandwidth simplifies the implementation (simpler filtering and timing recovery) at expense of channel bandwidth

Zero-excess-bandwidth pulse is unique \rightarrow ideal bandlimited pulses

non-zero excess bandwidth, pulse shape no longer unique

Commonly used pulses with nonzero excess bandwidth that satisfy the Nyquist criterion are the raised-cosine pulses, given by

$$g(t) = \left(rac{\sin(\pi t/T)}{\pi t/T}
ight) \left(rac{\cos(lpha \pi t/T)}{1 - (2lpha t/T)^2}
ight)$$

Fourier transforms of raised-cosine pulses:

$$G(f) = \begin{cases} T, & |f| \leq \frac{1-\alpha}{2T} \\ T\cos^2\left[\frac{\pi T}{2\alpha}\left(|f| - \frac{1-\alpha}{2T}\right)\right], & \frac{1-\alpha}{2T} < |f| \leq \frac{1+\alpha}{2T} \\ 0, & \frac{1+\alpha}{2T} < |f| \end{cases}$$

Refer to Fig 5.2

 $\alpha = \mathbf{0} \rightarrow \mathrm{ideally}$ bandlimited pulses

Other values of α , energy rolls of more gradually (α also roll-off factor)

Shape of roll-off is that of a cosine raised above abscissa.

Consider impact of a channel

Many important channels modeled as a linear time-invariant filter with impulse response b(t) and additive noise n(t)

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PAM signal applied to linear channel with impulse response b(t) and additive noise n(t):

$$r(t) = \int_{-\infty}^{\infty} b(\tau) \sum_{m=-\infty}^{\infty} a_m g(t - mT - \tau) d\tau + n(t)$$

rewritten as

$$r(t) = \sum_{m=-\infty}^{\infty} a_m h(t - mT) + n(t)$$

where h(t) = g(t) * b(t) is the convolution of g(t) with b(t):

$$h(t) = \int_{-\infty}^{\infty} g(\tau) b(t-\tau) d\tau$$

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 $r(t) \rightarrow$ received pulse \rightarrow also PAM if transmitted pulse is PAM:

- Different pulse shape
- added noise

Typical receiver front end consists of a receive filter f(t) followed by a sampler



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Receive filter perform several functions, including:

- compensating for the distortion of the channel
- diminishing the effect of additive noise

Receive filter conditions the received signal before sampling

If bandwidth of additive noise wider than that of transmitted signal, receive filter can reject out-of-band noise

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Receive filter might be chosen to avoid ISI after sampling

Output of receive filter (input to sampler):

$$y(t) = \sum_{m=-\infty}^{\infty} a_m p(t - mT) + n'(t)$$

where $p(t) = g(t) * b(t) * f(t) \rightarrow$ overall pulse shape

noise n'(t) is filtered version of the received noise n(t)

Receive filter output is another PAM signal, pulse shape p(t) and with added noise

To avoid ISI, overall pulse shape p(t) = g(t) * b(t) * f(t) must be Nyquist

Thus,
$$p(kT) = \delta_k$$
, or $\sum_m P(f - \frac{m}{T}) = T$

When this condition is satisfied, the k-th sample y(kT) reduces to a_k plus noise, with no interference from $\{a_{l\neq k}\}$

Since P(f) = G(f)B(f)F(f), a bandwidth limitation on the channel necessarily leads to the same bandwidth limitation on the overall pulse shape.

Thus, it is the channel bandwidth W that determines the maximum symbol rate, namely 1/T = 2W

5.1.3. ISI and Eye diagrams

Self study

Symbols independent and uniform from alphabet $\mathcal A$ of size $|\mathcal A|\to$ each symbol conveys $\mathsf{log}_2|\mathcal A|$ bits of information

Transmits 1/T symbols per second, bit rate is:

$$R_b = \frac{\log_2|\mathcal{A}|}{T} \mathsf{b}/\mathsf{s}$$

Increase bit rate

- Increase size of alphabet
- increase symbol rate

Symbol rate is bounded by the bandwidth constraints of channel

Size of alphabet constrained by:

- allowable transmitted power
- severity of the additive noise on the channel

Constraints on symbol rate and alphabet size limits available bit rate for a given channels

Spectral efficiency:

$$\nu = \frac{R_b}{W}$$

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Baseband PAM:

$$\nu = \frac{R_b}{W} = \frac{\log_2|\mathcal{A}|/T}{(1+\alpha)/(2T)} = \frac{2\log_2|\mathcal{A}|}{1+\alpha}$$

Maximal spectral efficiency:

$$\nu_{max} = 2\log_2|\mathcal{A}|$$

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5.2. Passband PAM

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Many practical communication channels are passband in nature \rightarrow frequency response that of bandpass filtered



Method 1

Start with suboptimal strategy \rightarrow pulse-amplitude-modulation double-sideband

Passband channel has bandwidth B

Start with a real-valued baseband PAM signal with bandwidth B/2

Modulate carrier frequency f_c , by multiplying f_c and baseband PAM:

$$s(t) = \sqrt{2}\cos(2\pi f_c t) \sum_k a_k g(t - kT)$$

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Method 1

Modulated signal will pass undistorted through channel when pulse shape g(t) is low-pass with bandwidth B/2

Avoiding ISI, symbol rate is twice the pulse shape bandwidth (Symbol rate = 1/T = B)

Maximal spectral efficiency of PAM-DSB with real alphabet $\mathcal A$ is $\mathsf{log}_2|\mathcal A|$

Method 2

Recognise upper sideband and lower sideband of s(t) conveys identical information

Double spectral efficiency by using single-sideband (SSB), transmit only one sideband

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 $\mathsf{Disadvantage} \to \mathsf{difficulty} \text{ in realizing filtering}$

Method 3

Recognise that PAM-DSB carries information only in in-phase component

Quadrature component is zero

Double the spectral efficiency of PAM-DSB by transmitting a second baseband PAM signal in quadrature:

$$s(t) = \sqrt{2}\cos(2\pi f_c t)\sum_k a'_k g(t-kT) - \sqrt{2}\sin(2\pi f_c t)\sum_k a^Q_k g(t-kT)$$

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Method 3

Bandwidth the same as PAM-DSB, but conveys twice as much information

Assume both baseband PAM signals use the same pulse shape

Symbols modulating in-phase and quadrature components are denoted $\{a_k^I\}$ and $\{a_k^Q\}$

 ${\rm QAM} \to \{a_k^I\}$ and $\{a_k^Q\}$ chosen independently from same real alphabet ${\cal A}$

See Fig 5.10

Complex Envelope

Can represent s(t) in terms of complex envelope:

$$s(t) = \sqrt{2} \operatorname{Re}\left\{\widetilde{s}(t) e^{j2\pi f_c t}\right\}$$

where the complex envelope of a passband PAM signal is:

$$\widetilde{s}(t) = \sum_{k} a_{k}g(t-kT)$$

with $a_k = a_k^I + j a_k^Q$

Complex Envelope

Observe that complex envelope of passband PAM looks exactly like real-valued baseband PAM signal

Passband PAM signal \rightarrow signal whose complex envelope is the baseband PAM signal with complex symbols and a real pulse shape

For a realization, refer to Fig 5-11 (Theoretical)

Comparing Method 2 and Method 3

Both passband PAM and PAM-SSB double spectral efficiency of PAM-DSB

 $\mathsf{PAM}\text{-}\mathsf{SSB} \to \mathsf{fixes}$ the bit rate but cuts bandwidth in half

passband $\mathsf{PAM} \to \mathsf{doubles}$ bit rate while keeping the bandwidth fixed

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Another representation for Passband PAM

Another presentation of passband PAM \rightarrow express data symbols a_m in polar coordinates

$$a_m = c_m e^{j \theta_m}$$

So that

$$s(t) = \sqrt{2} \operatorname{Re} \left\{ \sum_{-\infty}^{\infty} c_m e^{j2\pi f_c t + \theta_m} g(t - mT) \right\} \\ = \sqrt{2} \sum_{-\infty}^{\infty} c_m \cos(2\pi f_c t + \theta_m) g(t - mT)$$

Each pulse g(t - mT) multiplied by carrier, where amplitude and phase of the carrier is determined by the amplitude and phase of a_m

Sometimes called AM/PM

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Alphabet \rightarrow set \mathcal{A} of symbols available for transmission

Baseband signal has real-valued alphabet

Passband PAM signal \rightarrow alphabet that is a set of complex numbers

For real-valued and complex alphabets \rightarrow each symbol represents $\mathsf{log}_2|\mathcal{A}|$ bits

Complex-valued alphabet is best described by plotting the alphabet as a set of points in a complex plain

 $\mathsf{Plot} \to \mathsf{signal} \ \mathsf{constellation}$

Example 5-12 - on blackboard

Example 5-13 - on blackboard

Energy of alphabet:

Assumptions:

- All symbols are equally likely
- pulse shape is normalized to have unit energy

Expected energy E of a single passband PAM pulse transmitted in isolation,

$$s(t) = \sqrt{2} \operatorname{Re}\left\{\widetilde{s}(t) e^{j2\pi f_c t}
ight\}$$

with $\tilde{s}(t) = ag(t)$:

$$E = E[\int_{-\infty}^{\infty} s^{2}(t)dt]$$

= $E[\int_{-\infty}^{\infty} |\tilde{s}(t)|^{2}dt]$
= $E[|a|^{2}]\int_{-\infty}^{\infty} g(t)^{2}dt$
= $\frac{1}{|\mathcal{A}|}\sum_{a\in\mathcal{A}}|a|^{2}$

Power: Leave output

Alphabet Design

Distance between points in a constellation determines the likelihood that one point will be confused with another

Minimum distance d_{min} between two points key parameter of the constellation

Two constellations can be considered to have the approximately the same noise immunity if the minimum distance d_{min} is the same.

To make d_{min} the same for constellations with different number of points, higher point constellations require more transmit power

Either a power or an error-probability penalty associated with using larger constellations

Objective of signal constellation design \rightarrow maximize distance between symbols while not exceeding power constraint

Optimal constellations difficult to derive or costly to implement

Assume average power constraint

Performance of a constellation depends only on distances among symbols \rightarrow performance of constellation invariant under translation

Should translate a constellation so that its power is minimized

Power minimized if it has zero mean

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Given a set of symbols $\{a_i\}$, translate with complex number *m* such that the power

$$E[|a-m|^2] = \sum_{i=1}^{M} p_a(a_i)|a_i - m|^2$$

of translated symbol set $\{a_i - m\}$ is minimized

Best choice for translation: m = E[a]

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Proof:

for any other transformation n

$$E[|a - n|^2] = E[|(a - m) + (m - n)|^2]$$

= $E[|a - m|^2] + 2\text{Re}\{(m - n)^*(E[a] - m)\} + |m - n|^2$
= $E[|a - m|^2] + |m - n|^2$

Mean energy under translation n is larger than mean energy under translation m by $|m-n|^2$

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Problem of optimal design of constellation is complicated

QAM \rightarrow Square $\rightarrow M = 2^{b}$, b even \rightarrow Fig. 5-13

 $QAM \rightarrow M = 2^{b}$, b odd \rightarrow Fig. 5-14

PSK and PSK + amplitude modulation \rightarrow Fig. 5-15

Hexagonal constellations \rightarrow Fig 5-16 (Hexagonal refer to shape of decision regions)

<u>PSK</u>

- 2-PSK \rightarrow binary phase-shift keying (BPSK)
- $4\text{-PSK} \rightarrow \text{QPSK} \ / \ 4\text{-QAM}$

M elements of M-PSK:

$$a = \sqrt{E}e^{j2\pi m/M}, ext{ for } m \in \{0, \dots, M-1\}$$

 $\mathsf{Pure}\;\mathsf{PSK}\to\mathsf{constant}\;\mathsf{envelope}\to\mathsf{robust}\;\mathsf{against}\;\mathsf{amplifier}\;\mathsf{nonlinearities}$

5.2.3. Spectral Efficiency

Bit rate
$$R_b o \log_2 |\mathcal{A}| imes rac{1}{T}$$

Spectral efficiency $\nu = R_b/\text{bandwidth}$

Difference between baseband PAM and passband PAM is relationship between symbol rate and bandwidth

 $\mathsf{Passband} \to \mathsf{bandwidth} \ W \to \mathsf{maximal} \ \mathsf{symbol} \ \mathsf{rate} \ \mathsf{is} \ W$

Due to bandwidth of passband PAM signal being twice bandwidth of pulse shape (upconversion process)

$$\nu = \frac{R_b}{W} = \frac{\log_2|\mathcal{A}|}{1+\alpha}$$

5.2.3. Spectral Efficiency

Passband PAM lower spectral efficiency than baseband PAM baseband

Complex alphabet (Passband) is much larger than real alphabet (baseband)

If using QAM and transmit L levels on each of the two quadrature carriers:

$$\nu = \log_2 L^2 = 2 \cdot \log_2 L$$
 bits / sec-Hz

Same as for baseband PAM with L levels