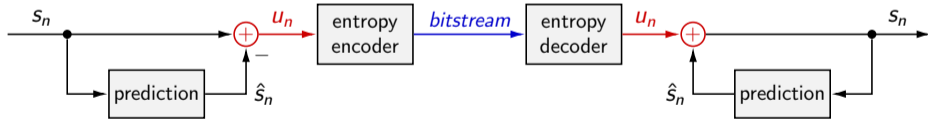


# Predictive Lossless Coding



## Last Lecture: Arithmetic Coding

### Arithmetic Coding

- Practical realization of Shannon-Fano-Elias Coding (using standard integer arithmetic)
- No codeword table, on-the-fly encoding and decoding

### Arithmetic Coding vs Huffman Coding

- For given block size  $N$ : Huffman coding is optimal, arithmetic coding is suboptimal
- Arithmetic coding is realizable for large  $N$ , while Huffman coding is not

### Coding Efficiency of Arithmetic Coding

- Given probabilities and large  $N$ : Coding efficiency is very close to theoretical optimum
- In practice: Coding efficiency depends on using suitable probabilities
  - ➔ Adaptive pmf estimation during encoding and decoding
  - ➔ Using conditional pmfs (switch adaptive pmfs during encoding and decoding)

# Arithmetic Coding with Adaptive Pmfs

## Basic Idea

- Update pmf after encoding of each symbol
- Update pmf after decoding of each symbol

## Straightforward realization

- Count occurrences  $N_k$  of alphabet letters  $a_k$
- $V$ -bit probabilities  $p_V(a_k)$  are given by

$$p_V(a_k) = \left\lfloor 2^V \cdot \frac{N_k}{\sum_k N_k} \right\rfloor \quad \left( \begin{array}{l} \text{rounding} \\ \text{down !} \end{array} \right)$$

- Initialization:  $\forall k, N_k = 1$

## Controlling Adaptation Speed

- One possibility:  
Rescale counts after sum exceeds some limit

```
class Pmf // example for adaptive pmf implementation
{
public:
    Pmf( int Vbits, int numLetters, int maxSumCounts )
        : V          ( Vbits )
          , maxSum    ( maxSumCounts ) // adaption speed
          , sumCounts( numLetters )
          , counts    ( numLetters, 1 ) // all counts = 1
    {}

    int operator[] ( int index ) const {
        return ( counts[ index ] << V ) / sumCounts;
    }

    void update( int index ) {
        counts[ index ]++;
        if( ++sumCounts >= maxSum ) {
            sumCounts = 0;
            for( auto& cnt : counts ) // rounding up !
                sumCounts += ( cnt = ( cnt + 1 ) >> 1 );
        }
    }

private:
    const int V; // number of bits for pmf
    const int maxSum; // maximum sum of counts
    int sumCounts; // sum of all counts
    vector<int> counts; // counts for letters
};
```

# Arithmetic Coding with Conditional Pmf's

## Basic Idea

- Switch pmf after encoding of each symbol
- Switch pmf after decoding of each symbol
- Can be combined with adaptive pmfs

## Conditions for 1D Signals (e.g., audio)

- Directly preceding sample
- Two (or more) directly preceding samples
- Function of one or more preceding samples

## Conditions for 2D Signals (e.g., images)

- One already coded neighboring sample
- Two or more already coded neighboring samples
- Function of already coded neighboring samples

```
void encodeMessage( const vector<int>& message, ... )
{
    vector<Pmf> pmfs( numLetters, {...} );
    ArithEncoder aenc( ... );
    int lastSymbol = 0;
    for( const auto& currSymbol : message ) {
        Pmf& currPmf = pmfs[ lastSymbol ];
        aenc.encode ( currSymbol, currPmf );
        currPmf.update( currSymbol );
        lastSymbol = currSymbol;
    }
    aenc.terminate();
}
```

```
vector<int> decodeMessage( int numSymbols, ... )
{
    vector<int> message;
    vector<Pmf> pmfs( numLetters, {...} );
    ArithDecoder adec( ... );
    int symbol = 0;
    while( message.size() < numSymbols ) {
        Pmf& currPmf = pmfs[ symbol ];
        symbol = adec.decode( currPmf );
        currPmf.update ( symbol );
        message.push_back( symbol );
    }
    return message;
}
```

# Example: Conditional Arithmetic Coding for Document Scans



- 1 **No condition:** 1 binary pmf  
→ File size = 418 369 bytes (39.74 %)
- 2 **Left neighbour:** 2 binary pmf's  
→ File size = 192 841 bytes (18.32 %)
- 3 **Left and above:** 4 binary pmf's  
→ File size = 120 198 bytes (11.42 %)
- 4 **Four neighbours:** 16 binary pmf's  
→ File size = 101 819 bytes (9.67 %)
- 5 **Eleven neighbours:** 2048 binary pmf's  
→ File size = 92 527 bytes (8.79 %)

Original:	1 052 713 bytes	
gzip:	183 705 bytes	(17.45 %)
bzip2:	169 049 bytes	(16.05 %)
lzip:	140 307 bytes	(13.33 %)

1098

PROCEEDINGS OF THE I.R.E.

September

## A Method for the Construction of Minimum-Redundancy Codes\*

DAVID A. HUFFMAN†, ASSOCIATE, IRE

*Summary*—An optimum method of coding an ensemble of messages consisting of a finite number of members is developed. A minimum-redundancy code is one constructed in such a way that the average number of coding digits per message is minimized.

### INTRODUCTION

ONE IMPORTANT METHOD of transmitting messages is to transmit in their place sequences of symbols. If there are more messages which might be sent than there are kinds of symbols available, then some of the messages must use more than one symbol. If it is assumed that each symbol requires the same time for transmission, then the time for transmission (length) of a message is directly proportional to the number of symbols associated with it. In this paper, the symbol or sequence of symbols associated with a given message will be called the "message code." The entire number of messages which might be transmitted will be called the "message ensemble." The mutual agreement between the transmitter and the receiver about the meaning of the code for each message of the ensemble will be called the "ensemble code."

Probably the most familiar ensemble code was stated in the phrase "one if by land and two if by sea." In this

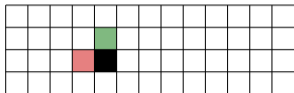
will be defined here as an ensemble code which, for a message ensemble consisting of a finite number of members,  $N$ , and for a given number of coding digits,  $D$ , yields the lowest possible average message length. In order to avoid the use of the lengthy term "minimum-redundancy," this term will be replaced here by "optimum." It will be understood then that, in this paper, "optimum code" means "minimum-redundancy code."

The following basic restrictions will be imposed on an ensemble code:

- (a) No two messages will consist of identical arrangements of coding digits.
- (b) The message codes will be constructed in such a way that no additional indication is necessary to specify where a message code begins and ends once the starting point of a sequence of messages is known.

Restriction (b) necessitates that no message be coded in such a way that its code appears, digit for digit, as the first part of any message code of greater length. Thus, 01, 102, 111, and 202 are valid message codes for an ensemble of four members. For instance, a sequence of these messages 1111022020101111102 can be broken up

## Example: Conditional Arithmetic Coding for 8-bit Images



Original:	262 159 bytes	(512 × 512)
gzip:	222 999 bytes	(85.06 %)
bzip2:	173 877 bytes	(66.33 %)
lzip:	180 000 bytes	(68.66 %)

### 1 No condition:

- 256 probability masses ( $2^8$ )
- File size = 240 112 bytes (91.59 %)

### 2 Left sample:

- 65 536 probability masses ( $2^8 \cdot 2^8$ )
- File size = 179 179 bytes (68.35 %)

### 3 Left sample and above sample:

- 16 777 216 probability masses ( $2^8 \cdot 2^8 \cdot 2^8$ )
- File size = 221 849 bytes (84.62 %)

- **Too many probability masses**
- **Pmfs do not adapt to image statistics**



## Example: Conditional Arithmetic Coding for 16-bit Audio



### 1 No condition:

- 65 536 probability masses ( $2^{16}$ )
- File size = 23 700 606 bytes (89.23 %)

### 2 Previous sample:

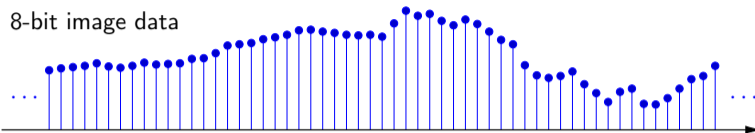
- 4 294 967 296 probability masses ( $2^{16} \cdot 2^{16}$ )
  - Not possible on normal computers
  - Requires 16 GByte of memory  
(when we use 32 bit per probability)
  - Would not adapt well to statistics
- Cannot exploit dependencies between symbols using this type of coding !!!

Original:	26 559 960 bytes	(5:01 minutes)
gzip:	24 926 843 bytes	(93.85 %)
bzip2:	22 445 509 bytes	(84.51 %)
lzip:	23 777 258 bytes	(89.52 %)



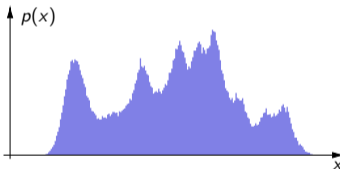
# Analysis: Typical Properties of Real Signals

8-bit image data



## Marginal Pmf

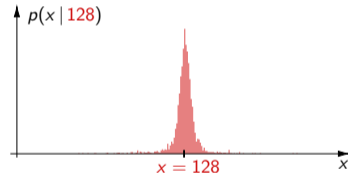
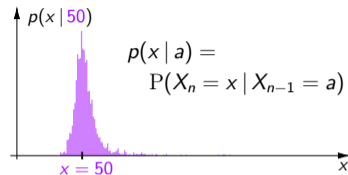
- Not much room for compression
- Entropy  $H = 7.45$  (8-bit data)



## Conditional Pmfs

- Significantly narrower than marginal pmf (room for compression)
- Shape is very similar for all conditions, but shifted by  $x_{n-1}$

→ Idea: Code difference to previous sample !!!





# Simple Predictive Coding using Preceding Sample



## Encoding

- Generate difference sample

$$u_n = s_n - s_{n-1}$$

- Encode difference sample  $u_n$

## Decoding

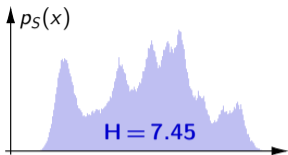
- Decode difference sample  $u_n$
- Reconstruct original sample

$$s_n = s_{n-1} + u_n$$

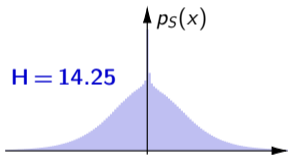
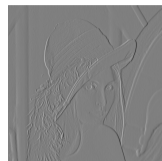
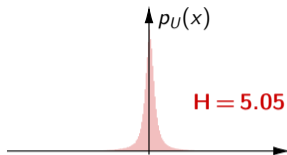
## Basic Effect

- Prediction removes large part of inter-symbol dependencies before actual entropy coding
- Reduces required complexity for the entropy coding (e.g., marginal instead of conditional coding)

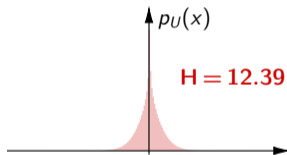
# How Well Does That Type of Prediction Work ?



$$u_n = s_n - s_{n-1}$$

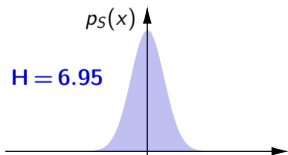


$$u_n = s_n - s_{n-1}$$

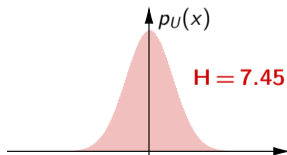


saves nearly  
2 bits per  
sample !

White  
Gaussian  
Noise



$$u_n = s_n - s_{n-1}$$



prediction  
increases  
entropy !

# General Predictive Coding



## Predictive Lossless Coding

- Predict current sample  $s_n$  using a function of preceding samples

$$\hat{s}_n = f(s_{n-1}, s_{n-2}, \dots)$$

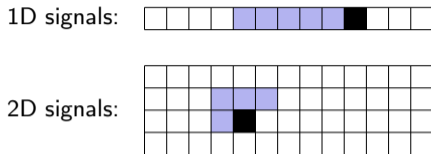
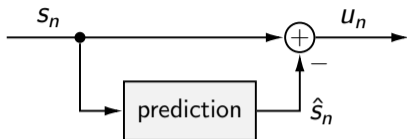
- Entropy coding of prediction error samples

$$u_n = s_n - \hat{s}_n$$

- Decoder uses exactly the same prediction and reconstructs the original samples according to

$$s_n = \hat{s}_n + u_n$$

# Choice of Observation Set and Predictor



## Choice of Observation Set $\mathcal{B}_n$

- Use small number of preceding samples for prediction
- Choose the samples  $\mathcal{B}_n$  with highest dependencies to current sample  $s_n$ 
  - 1D signals (e.g., audio): N directly preceding samples  $\rightarrow \mathcal{B}_n = \{s_{n-1}, s_{n-2}, \dots, s_{n-N}\}$
  - 2D signals (e.g., images): Samples in causal direct neighborhood

## Choice of Predictor for given Observation Set

- Question: What function  $\hat{s}_n = f(\mathcal{B}_n)$  should we use for prediction?
- Want to minimize entropy  $H(U_n)$  of prediction error samples  $\{u_n\}$  (difficult to access directly)
- Can minimize variance  $\sigma_U^2$  and ignore shape of pmf  $p(u)$

# Optimization Criterion for Deriving Predictor

## Typical Optimization Criterion

- Minimize Energy of Prediction Error Signal

$$\varepsilon_U^2 = \mathbb{E}\{U_n^2\} = \mathbb{E}\left\{\left(S_n - \hat{S}_n\right)^2\right\} = \mathbb{E}\left\{\left(S_n - f(\mathcal{B}_n)\right)^2\right\}$$

- Reformulate prediction error energy

$$\begin{aligned}\varepsilon_U^2 &= \mathbb{E}\{U_n^2\} = \mathbb{E}\left\{(U_n - \mu_U + \mu_U)^2\right\} \\ &= \mathbb{E}\left\{(U_n - \mu_U)^2\right\} + \mu_U^2 + 2\mu_U \mathbb{E}\{U_n - \mu_U\} \\ &= \sigma_U^2 + \mu_U^2\end{aligned}$$

- ➔ Minimization of squared prediction error  $\varepsilon_U^2$  implies minimization of variance  $\sigma_U^2$  and mean  $\mu_U^2$
- ➔ Minimize the width of the pmf  $p_U(u)$ , but ignore its actual shape
- ➔ Suitable alternative to minimization of the entropy  $H(U)$

## Optimal Predictor for Given Observation Set

- Question: What value  $a$  minimizes the prediction error energy ?

$$\begin{aligned}
 \varepsilon_U^2 &= \mathbb{E}\{ (S_n - a)^2 \} = \mathbb{E}\{ (S_n - \mathbb{E}\{ S_n \} + \mathbb{E}\{ S_n \} - a)^2 \} \\
 &= \mathbb{E}\{ (S_n - \mathbb{E}\{ S_n \})^2 \} + \mathbb{E}\{ (\mathbb{E}\{ S_n \} - a)^2 \} + 2 \mathbb{E}\{ (S_n - \mathbb{E}\{ S_n \})(\mathbb{E}\{ S_n \} - a) \} \\
 &= \sigma_S^2 + \underbrace{(\mathbb{E}\{ S_n \} - a)^2}_{\geq 0}
 \end{aligned}$$

- ➔ Prediction error energy is minimized by mean, i.e., by setting  $a = \mathbb{E}\{ S_n \}$

### Minimization of mean-squared prediction error for given observation set $\mathcal{B}_n$

- Similar optimization problem: Minimize  $\mathbb{E}\{ (S_n - f(\mathcal{B}_n))^2 \mid \mathcal{B}_n \}$
- ➔ Solution: **Optimal predictor is given by the conditional mean**

$$\hat{s}_n = f_{\text{opt}}(\mathcal{B}_n) = \mathbb{E}\{ S_n \mid \mathcal{B}_n \}$$

- ➔ General case requires storage of large tables (often impractical and complex)

# Optimal Predictor for Autoregressive Sources

## Autoregressive Sources of Order $m$

- Good probabilistic model for many real signals: **AR(m) process**

$$S_n = \mu_s + \sum_{k=1}^m a_k \cdot (S_{n-k} - \mu_s) + Z_n \quad (Z_n \text{ is zero-mean iid process})$$

- ➔ Optimal predictor if we know the entire past, i.e.,  $\mathcal{B}_n = \{s_{n-1}, s_{n-2}, \dots\}$

$$\begin{aligned} \mathbb{E}\{S_n | \mathcal{B}_n\} &= \mathbb{E}\left\{ \mu_s + \sum_{k=1}^m a_k \cdot (S_{n-k} - \mu_s) + Z_n \mid s_{n-1}, s_{n-2}, \dots \right\} \\ &= \mu_s \left( 1 - \sum_{k=1}^m a_k \right) + \sum_{k=1}^m a_k \cdot \mathbb{E}\{S_{n-k} \mid s_{n-1}, s_{n-2}, \dots\} + \mathbb{E}\{Z_n \mid s_{n-1}, s_{n-2}, \dots\} \\ &= a_0 + \sum_{k=1}^m a_k \cdot s_{n-k} \quad \text{with} \quad a_0 = \mu_s \left( 1 - \sum_{k=1}^m a_k \right) \end{aligned}$$

- ➔ **Optimal predictor is an affine function of the past  $m$  samples**

## Variance and Mean for Affine Prediction

Affine Predictor: 
$$\hat{s}_n = a_0 + \sum_{k=1}^K a_k \cdot b_k$$
 for any observation set  $\mathcal{B}_n = \{b_1, b_2, \dots, b_K\}$

### ■ Mean $\mu_U$ of prediction error

$$\begin{aligned} \mu_U^2 = \mathbb{E}\{U_n\} &= \mathbb{E}\left\{S_n - a_0 - \sum_{k=1}^K a_k B_k\right\} \\ &= \mu_S \left(1 - \sum_{k=1}^K a_k\right) - a_0 \end{aligned}$$

→ Can be forced to  $\mu_U = 0$  by setting  $a_0 = \mu_S \left(1 - \sum_{k=1}^K a_k\right)$

### ■ Variance $\sigma_U^2$ of predictor error does not depend on constant offset $a_0$

$$\begin{aligned} \sigma_U^2 &= \mathbb{E}\left\{(U_n - \mathbb{E}\{U_n\})^2\right\} = \mathbb{E}\left\{\left(S_n - a_0 - \sum_{k=1}^K a_k B_k - \mathbb{E}\left\{S_n - a_0 - \sum_{k=1}^K a_k B_k\right\}\right)^2\right\} \\ &= \mathbb{E}\left\{\left(\left(S_n - \sum_{k=1}^K a_k B_k\right) - \mu_S \left(1 - \sum_{k=1}^K a_k\right)\right)^2\right\} \neq f(a_0) \end{aligned}$$



# Affine and Linear Prediction

Affine Predictor:

$$\hat{s}_n = a_0 + \sum_{k=1}^K a_k \cdot b_k$$

for any observation set  $\mathcal{B}_n = \{b_1, b_2, \dots, b_K\}$

→ For a minimization of the prediction error variance  $\sigma_U^2$ , a linear predictor is sufficient

$$\hat{s}_n = \sum_{k=1}^K a_k \cdot b_k$$

→ The mean  $\mu_U$  of the prediction error can be forced to zero by additionally setting

$$a_0 = \mu_S \left( 1 - \sum_{k=1}^K a_k \right)$$

→ **How can we derive optimal prediction parameters  $\{a_1, a_2, \dots, a_K\}$  for a given observation set and a given source ?**

## Consider: Linear Prediction and Minimization of Variance

### Vector Notation (for simplifying derivations)

- Observation set:  $\mathbf{B}_n = (B_1, B_2, \dots, B_K)^T$
- Prediction parameters:  $\mathbf{a} = (a_1, a_2, \dots, a_K)^T$
- ➔ Linear predictor:  $\hat{S}_n = \mathbf{a}^T \cdot \mathbf{B}_n$
- ➔ Predictor error:  $U_n = S_n - \hat{S}_n = S_n - \mathbf{a}^T \mathbf{B}_n$

### Optimization Problem

- Minimization of prediction error variance

$$\begin{aligned} \sigma_U^2(\mathbf{a}) &= \mathbb{E} \left\{ \left( U_n - \mathbb{E}\{U_n\} \right)^2 \right\} = \mathbb{E} \left\{ \left( S_n - \mathbf{a}^T \mathbf{B}_n - \mathbb{E}\{S_n - \mathbf{a}^T \mathbf{B}_n\} \right)^2 \right\} \\ &= \mathbb{E} \left\{ \left( \left( S_n - \mathbb{E}\{S_n\} \right) - \mathbf{a}^T \left( \mathbf{B}_n - \mathbb{E}\{\mathbf{B}_n\} \right) \right)^2 \right\} \end{aligned}$$

## Reformulate Prediction Error Variance

### ■ Prediction error variance

$$\begin{aligned}
 \sigma_U^2(\mathbf{a}) &= \mathbb{E} \left\{ \left( (S_n - \mathbb{E}\{S_n\}) - \mathbf{a}^T (\mathbf{B}_n - \mathbb{E}\{\mathbf{B}_n\}) \right)^2 \right\} \\
 &= \mathbb{E} \left\{ (S_n - \mathbb{E}\{S_n\})^2 \right\} - 2 \mathbf{a}^T \cdot \mathbb{E} \left\{ (S_n - \mathbb{E}\{S_n\}) (\mathbf{B}_n - \mathbb{E}\{\mathbf{B}_n\}) \right\} \\
 &\quad + \mathbf{a}^T \cdot \mathbb{E} \left\{ (\mathbf{B}_n - \mathbb{E}\{\mathbf{B}_n\}) (\mathbf{B}_n - \mathbb{E}\{\mathbf{B}_n\})^T \right\} \cdot \mathbf{a} \\
 &= \sigma_S^2 - 2 \mathbf{a}^T \mathbf{c} + \mathbf{a}^T \mathbf{C}_B \mathbf{a}
 \end{aligned}$$

with  $\mathbf{C}_B$  and  $\mathbf{c}$  being given by

$$\begin{aligned}
 \mathbf{C}_B &= \mathbb{E} \left\{ (\mathbf{B}_n - \mathbb{E}\{\mathbf{B}_n\}) (\mathbf{B}_n - \mathbb{E}\{\mathbf{B}_n\})^T \right\} \\
 \mathbf{c} &= \mathbb{E} \left\{ (S_n - \mathbb{E}\{S_n\}) (\mathbf{B}_n - \mathbb{E}\{\mathbf{B}_n\}) \right\}
 \end{aligned}$$

## Auto-Covariance Matrix of the Observation Set

$$\mathbf{C}_B = \mathbb{E} \left\{ \left( \mathbf{B}_n - \mathbb{E} \{ \mathbf{B}_n \} \right) \left( \mathbf{B}_n - \mathbb{E} \{ \mathbf{B}_n \} \right)^T \right\} = \sigma_S^2 \cdot \begin{bmatrix} 1 & \varrho_{1,2} & \varrho_{1,3} & \cdots & \varrho_{1,K} \\ \varrho_{2,1} & 1 & \varrho_{2,3} & \cdots & \varrho_{2,K} \\ \varrho_{3,1} & \varrho_{3,2} & 1 & \cdots & \varrho_{3,K} \\ \vdots & \vdots & \vdots & \ddots & \vdots \\ \varrho_{K,1} & \varrho_{K,2} & \varrho_{K,3} & \cdots & 1 \end{bmatrix}$$

- Each entry represents the correlation coefficient between two samples of the observation set

$$\varrho_{a,b} = \varrho_{b,a} = \frac{\text{cov}(B_a, B_b)}{\sigma_S^2} = \frac{\mathbb{E} \left\{ \left( B_a - \mu_S \right) \left( B_b - \mu_S \right) \right\}}{\mathbb{E} \left\{ \left( S_n - \mu_S \right)^2 \right\}}$$

- Property of the source (and choice of observation set)
- Can be measured for given signal (or set of signals)

## Cross-Covariance Vector of the Observation Set and Current Sample

$$\mathbf{c} = \mathbb{E} \left\{ \left( S_n - \mathbb{E} \{ S_n \} \right) \left( \mathbf{B}_n - \mathbb{E} \{ \mathbf{B}_n \} \right) \right\} = \sigma_S^2 \cdot \begin{bmatrix} \rho_{c,1} \\ \rho_{c,2} \\ \rho_{c,3} \\ \vdots \\ \rho_{c,K} \end{bmatrix}$$

- Each entry represents the correlation coefficient between the sample to be predicted and a sample of the observation set

$$\rho_{c,k} = \frac{\text{cov}(S_n, B_k)}{\sigma_S^2} = \frac{\mathbb{E} \left\{ \left( S_n - \mu_S \right) \left( B_k - \mu_S \right) \right\}}{\mathbb{E} \left\{ \left( S_n - \mu_S \right)^2 \right\}}$$

- Property of the source (and choice of observation set)
- Can be measured for given signal (or set of signals)

# Optimal Linear and Affine Prediction

- Goal: Minimization of prediction error variance  $\sigma_U^2$

$$\sigma_U^2(\mathbf{a}) = \sigma_S^2 - 2\mathbf{a}^T \mathbf{c} + \mathbf{a}^T \mathbf{C}_B \mathbf{a}$$

- Set derivative with respect to  $\mathbf{a}$  equal to zero

$$\frac{\partial}{\partial \mathbf{a}} \sigma_U^2(\mathbf{a}) = -2\mathbf{c} + 2\mathbf{C}_B \cdot \mathbf{a} = 0$$

- **Yule-Walker equations** (linear equation system)

optimal predictor  $\mathbf{a}$  solves

$$\mathbf{C}_B \cdot \mathbf{a} = \mathbf{c}$$

- Optimal affine prediction: Additionally, set

$$a_0 = \mu_S \left( 1 - \sum_{k=1}^K a_k \right)$$

## Summary: Derivation of Optimal Affine Predictor for Given Source

- 1 Choose suitable observation set  $\mathcal{B}_n = \{b_1, b_2, \dots, b_K\}$
- 2 Determine mean  $\mu_S$  of sources and required correlation coefficients  $\varrho_{k,\ell}$
- 3 Solve linear equation system for determining prediction parameters  $a_1, a_2, \dots, a_K$

$$\begin{bmatrix} 1 & \varrho_{1,2} & \varrho_{1,3} & \cdots & \varrho_{1,K} \\ \varrho_{2,1} & 1 & \varrho_{2,3} & \cdots & \varrho_{2,K} \\ \varrho_{3,1} & \varrho_{3,2} & 1 & \cdots & \varrho_{3,K} \\ \vdots & \vdots & \vdots & \ddots & \vdots \\ \varrho_{K,1} & \varrho_{K,2} & \varrho_{K,3} & \cdots & 1 \end{bmatrix} \cdot \begin{bmatrix} a_1 \\ a_2 \\ a_3 \\ \vdots \\ a_K \end{bmatrix} = \begin{bmatrix} \varrho_{c,1} \\ \varrho_{c,2} \\ \varrho_{c,3} \\ \vdots \\ \varrho_{c,K} \end{bmatrix}$$

- 4 Determine constant offset

$$a_0 = \mu_S \left( 1 - \sum_{k=1}^K a_k \right)$$

## Example: Prediction of Audio Signals using Three Preceding Samples

- 1 Choice of observation set for sample  $s_n$ :  $\mathcal{B}_n = \{s_{n-1}, s_{n-2}, s_{n-3}\}$
- 2 Need to determine mean  $\mu_S$  and three correlation coefficients  $\rho_k$  (for  $k = 1, 2, 3$ )

$$\mu_S = E\{S_n\} = \frac{1}{N} \sum_{n=1}^N s_n$$

$$\rho_k = \frac{E\{(S_n - \mu_S)(S_{n-k} - \mu_S)\}}{E\{(S_n - \mu_S)^2\}} = \frac{\sum_{n=k}^N (s_n - \mu_S)(s_{n-k} - \mu_S)}{\sum_{n=k}^N (s_n - \mu_S)^2}$$

### Example



$$\mu_S = -37.6917$$

$$\rho_1 = 0.9581$$

$$\rho_2 = 0.8619$$

$$\rho_3 = 0.7564$$



## Example: Prediction of Audio Signals using Three Preceding Samples

### 3 Solve linear equation system

$$\begin{bmatrix} 1 & \varrho_1 & \varrho_2 \\ \varrho_1 & 1 & \varrho_1 \\ \varrho_2 & \varrho_1 & 1 \end{bmatrix} \cdot \begin{bmatrix} a_1 \\ a_2 \\ a_3 \end{bmatrix} = \begin{bmatrix} \varrho_1 \\ \varrho_2 \\ \varrho_3 \end{bmatrix} \quad \Longrightarrow \quad \begin{aligned} a_1 &= 1.9409 \\ a_2 &= -1.4580 \\ a_3 &= 0.4804 \end{aligned}$$

### 4 Determine constant offset

$$a_0 = \mu_S \left( 1 - \sum_{k=1}^K a_k \right) \quad \Longrightarrow \quad a_0 = -1.3833$$

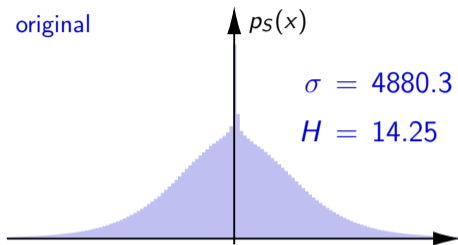
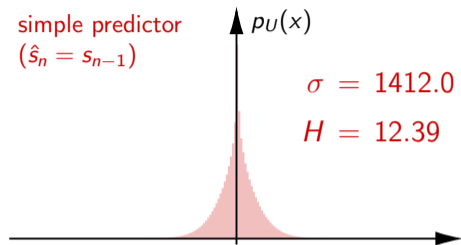
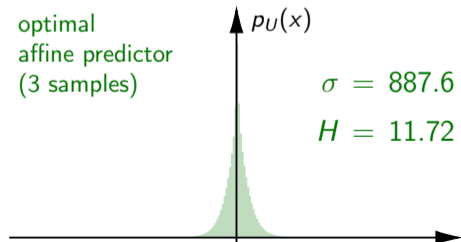
### → Predictor is given by

$$\begin{aligned} \hat{s}_n &= \text{round} \left( a_0 + a_1 \cdot s_{n-1} + a_2 \cdot s_{n-2} + a_3 \cdot s_{n-3} \right) \\ &= \text{round} \left( -1.3833 + 1.9409 \cdot s_{n-1} - 1.4580 \cdot s_{n-2} + 0.4804 \cdot s_{n-3} \right) \end{aligned}$$

**Note: Predictor must be rounded to an integer** (want to apply lossless coding)

# Example: Prediction of Audio Signals using Three Preceding Samples

original

simple predictor  
( $\hat{s}_n = s_{n-1}$ )optimal  
affine predictor  
(3 samples)

# Potential Improvements for Audio Coding

## 1 Use more samples as observation set

- May improve the affine predictor
- Requires transmission of more prediction parameters
- Audio compression format **FLAC** uses up to 32 preceding samples

## 2 Adapt predictor during coding

- Audio signals have instationary properties
- A single predictor may not be suitable for all parts of an audio stream
- Split data into chunks and determine best predictor for each chunk

# Example: Prediction of Image Signals using Three Neighboring Samples

## 1 Choice of prediction structure

$$\hat{X} = a_0 + a_L \cdot L + a_A \cdot A + a_C \cdot C$$

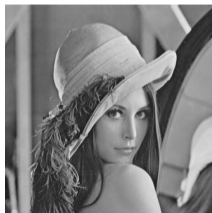
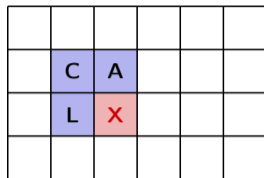
## 2 Determine mean and required correlation factors

$$\rho_{\text{hor}} = \frac{\sum_{x,y} (s[x,y] - \mu_S)(s[x-1,y] - \mu_S)}{\sum_{x,y} (s[x,y] - \mu_S)^2}$$

$$\rho_{\text{ver}} = \frac{\sum_{x,y} (s[x,y] - \mu_S)(s[x,y-1] - \mu_S)}{\sum_{x,y} (s[x,y] - \mu_S)^2}$$

$$\rho_{\text{al}} = \frac{\sum_{x,y} (s[x,y] - \mu_S)(s[x-1,y-1] - \mu_S)}{\sum_{x,y} (s[x,y] - \mu_S)^2}$$

$$\rho_{\text{ar}} = \frac{\sum_{x,y} (s[x,y] - \mu_S)(s[x+1,y-1] - \mu_S)}{\sum_{x,y} (s[x,y] - \mu_S)^2}$$



$$\begin{aligned} \mu_S &= 124.05 \\ \rho_{\text{hor}} &= 0.9722 \\ \rho_{\text{ver}} &= 0.9850 \\ \rho_{\text{al}} &= 0.9598 \\ \rho_{\text{ar}} &= 0.9689 \end{aligned}$$

## Example: Prediction of Image Signals using Three Neighboring Samples

### 3 Solve linear equation system

$$\begin{bmatrix} 1 & \varrho_{\text{ar}} & \varrho_{\text{ver}} \\ \varrho_{\text{ar}} & 1 & \varrho_{\text{hor}} \\ \varrho_{\text{ver}} & \varrho_{\text{hor}} & 1 \end{bmatrix} \cdot \begin{bmatrix} a_L \\ a_A \\ a_C \end{bmatrix} = \begin{bmatrix} \varrho_{\text{hor}} \\ \varrho_{\text{ver}} \\ \varrho_{\text{al}} \end{bmatrix} \implies \begin{aligned} a_L &= 0.5892 \\ a_A &= 0.8255 \\ a_C &= -0.4262 \end{aligned}$$

	C	A			
	L	X			

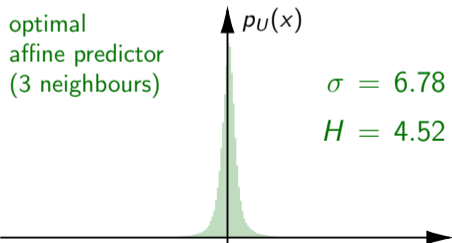
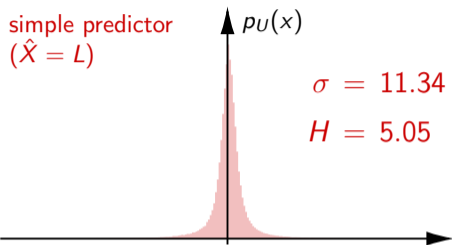
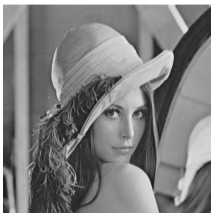
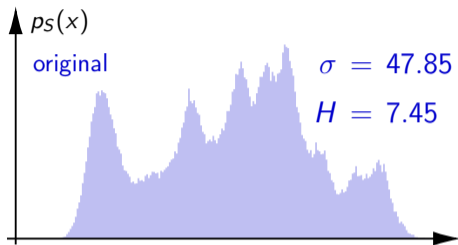
### 4 Determine constant offset

$$a_0 = \mu_S (1 - a_L - a_A - a_C) \implies a_0 = 1.4198$$

### → Predictor is given by

$$\begin{aligned} \hat{X} &= \text{round} \left( a_0 + a_L \cdot L + a_A \cdot A + a_C \cdot C \right) \\ &= \text{round} \left( 1.4198 + 0.5892 \cdot L + 0.8255 \cdot A - 0.4262 \cdot C \right) \end{aligned}$$

# Example: Prediction of Image Signals using Three Neighboring Samples



# Potential Improvements for Image Coding

## 1 Use more samples as observation set

- May improve the affine predictor
- Requires transmission of more prediction parameters

## 2 Adapt predictor during coding

- Image signals have instationary properties: Direction of edges plays an important role
- Split image into blocks and choose predictor for each block
- May be sufficient to choose between pre-defined predictors
  - ▶ Horizontal predictor for parts with horizontal edges
  - ▶ Vertical predictor for parts with vertical edges
  - ▶ Some further predictors

## 3 Non-linear Predictors

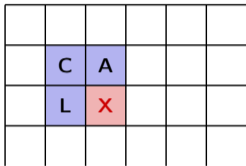
- Non-linear predictors may be able to deal with different edge directions

## Examples for Non-Linear Predictors

### LOCO Predictor used in JPEG-LS

- Each sample  $X$  is predicted according to

$$\hat{X} = \begin{cases} \min(L, A) & : C \geq \max(L, A) \\ \max(L, A) & : C \leq \min(L, A) \\ L + A - C & : \text{otherwise} \end{cases}$$

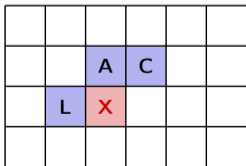


### Motion Vector Prediction in Video Coding Standards

- Motion vector  $\mathbf{m}$  is predicted by component-wise median

$$\hat{m}_x(X) = \text{median}(m_x(A), m_x(B), m_x(C))$$

$$\hat{m}_y(X) = \text{median}(m_y(A), m_y(B), m_y(C))$$





# Summary of Lecture

## Predictive Lossless Coding

- Entropy coding of prediction error signals  $u_n = s_n - \hat{s}_n$
- Simple and effective way to exploit dependencies between neighbouring samples
- Complexity reduction relative to more general conditional entropy coding

## Optimal Prediction

- Given by conditional mean for an observation set
- Complex due to requirement of large tables (similar to conditional entropy coding)

## Affine and Linear Prediction

- Simple structure of predictor, low-complex implementations possible
- Optimal prediction parameters are given by solution of Yule-Walker equations
- For instationary sources (such as audio, image, video signals):
  - Determine predictor for smaller sets of samples (still large enough)
  - Determine optimal predictor or choose between set of predefined predictors

# Exercise: Lossless Image Compression Challenge (Part I)

## Implement an encoder and decoder for lossless coding of 8-bit color images:

- 1 We use the PPM format as raw data format:
  - The encoder should read the original images in PPM format.
  - The decoder should write the reconstructed images in PPM format.

Example images (24 PPM images of the Kodak set) are provided on the course web-site (and in the KVV).

- 2 Use coding techniques that you learned for efficiently compressing the 8-bit color images.
  - A combination of prediction and entropy coding of the prediction errors is suggested.
  - Start with a simple (but working) approach and try to improve your codec step by step.

### structure of “ppm” files:

```
P6           // ascii (fixed)
width height // ascii
255         // ascii (max. value)
<binary data> // binary
```

### binary data:

- pixels in raster-scan order (line by line)
- each pixel consists of three 8-bit values
  - ➔ R: red component (0..255)
  - ➔ G: green component (0..255)
  - ➔ B: blue component (0..255)
- the values R, G, B for a pixel follow each other (before the values for the next pixel)

### suggestion:

- Store the red, green, and blue components of an image into separate arrays
- Code the color components independently