
JOINT SOURCE-CHANNEL TURBO TECHNIQUES FOR WIRELESS MULTIMEDIA COMMUNICATION

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Outline

- Applications and problems addressed
- Soft-decision source decoding
- Source-channel turbo coding/decoding
- Source codes with properties for source-channel coding
- Source codes with inherent resilience
- Conclusion

Outline

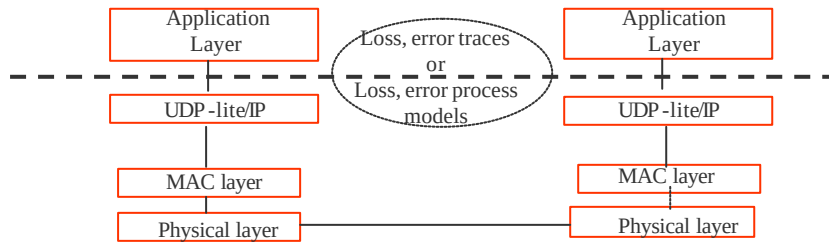
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Application Domains (1)

- Communication over networks supporting transparent modes, e.g.:
 - In the MAC layer: 3GPP, Bluetooth, ...
 - In the transport layer: UDP-lite, DCCP
- Digital media storage (DVD, CD):
 - Trend to relax the drastic residual error rate (10^{-20} => up to 25% redundancy)
 - Reducing the need for error correcting codes (hence increasing the storage capacity) while limiting effects of errors on quality
- Transmissions not relying on IP, e.g.:
 - Broadcasting applications such as Digital Radio Mondiale

Application domains (2): Example

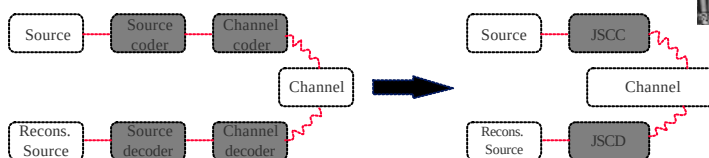
- Communication with transparent modes to save bandwidth



	BSC		3GPP		WIFI	
	UDP	UDP-Lite	UDP	UDP-Lite	UDP	UDP-Lite
mean number of received packets	6422	9466	9751	9887	9112	9742
% mean gain		47,4		1,4		6,9

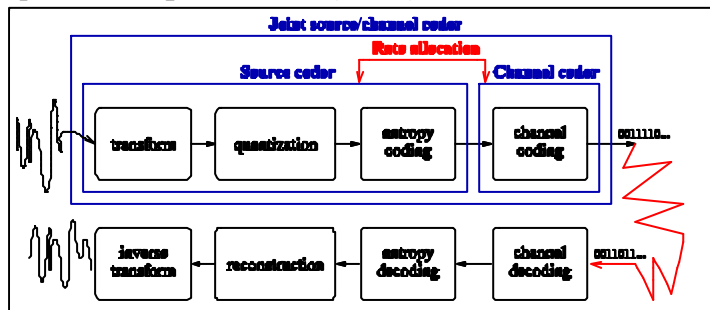
Introduction to the problem (1)

- Optimize the end-to-end QoS on wireless links
 - Narrow bands
 - Transmission noise (errors, erasures)
- In the traditional set-up
 - Source coding does not care about channel errors
 - High sensitivity of VLCs to errors
 - Decoder de-synchronisation problem
 - Use of error correcting codes



Introduction to the problem (2)

- Channel coding (ECC) is efficient against noise
 - But channel characteristics must be known at the time of encoding and may vary in time => residual bit errors
 - The separation theorem does not account for practical constraints (bounded delay, complexity)
- Unequal error protection (UEP)



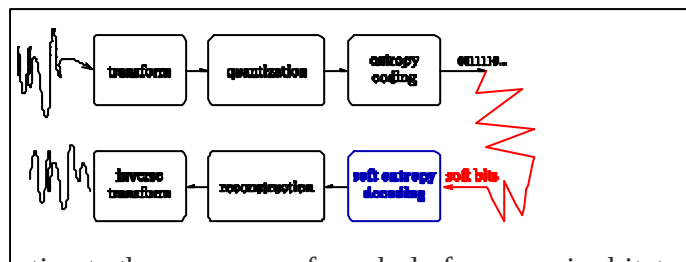
Introduction to the problem (3)

- Self-synchronization of VLCs has been extensively studied, e.g. [Maxted & al. 85, Zhou & al. 02]
 - “self-synchronization”: in terms of Levenshtein distance (not strict sense synchronization)
- Design of Reversible Variable Length Codes (RVLCs) [Takishima & al. 95]
- Robust decoding methods exploiting the sub-optimality of the source code
 - Bayesian Estimation => Soft-Decision Source Decoding [Balakirsky 97]
 - Turbo-VLC [Hagenauer00, Guyader, Fabre & Guillemot 01, ...]

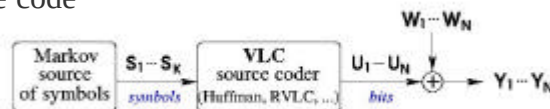
Outline

- Applications and problems addressed
- Soft decoding of VLCs
 - Tree codes (e.g., Huffman)
 - Soft synchronisation
 - Arithmetic codes
 - Trellis and complexity issues
 - Application to JPEG-2000 and to H.264/MPEG-4 AVC
- Source-channel turbo coding/decoding
- Source codes with properties for source-channel coding
- Source codes with inherent resilience features
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Soft Decoding of VLCs



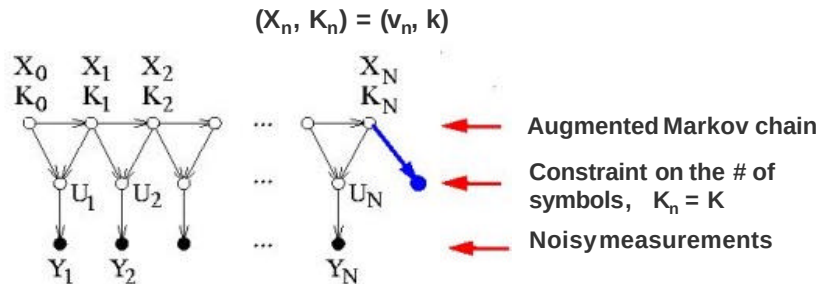
Goal: - To estimate the sequence of symbols from a noisy bitstream by exploiting the correlation in the chain and the sub-optimality of the code



=> Modelling the dependencies between the processes of the chain with a **Bayesian network formalism**

Soft Decoding of VLCs

- To force an additional termination constraint



=> **Bit/symbol trellis [Hagenauer 00]**

- Optimum performance but complexity growing as a quadratic function of the sequence length

Soft Decoding of VLCs

- Accounting for symbol correlation (**source with memory**)
 - Markov source + VLC coding process: Product Model

$$(X_n, K_n) = (s, v, k)$$

↑ The model keeps track of the last symbol produced

=> **Augmented bit/symbol trellis**

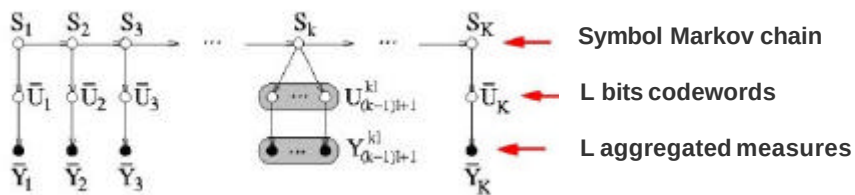
- Tree-structure graph amenable to fast estimation algorithms (BCJR, Viterbi)
- But state space of high dimension (untractable in practice)
- One can instead proceed in 2 steps [Guyader, Fabre & Guillemot 01]

Soft Decoding of VLCs

- 1- Bit clock : estimate $(X_n, K_n) = (v, k)$ assuming independent symbols, exploiting the inner codeword correlation
 - $\Rightarrow$ State space of reduced dimension
 - $\Rightarrow$ But sub-optimal (inter-symbol correlation not exploited)
- 2- To exploit the inter-symbol correlation via an estimation run on the Markov model of the sequence of symbols
 - $\Rightarrow$ Need for a symbol clock Markov model of the sequence of symbols

Soft Decoding of VLCs

- **Symbol clock model:** the simple case of FLCs:



- Fast estimation algorithms - MAP or MPM – on (S_k)

$$\hat{S}_k = \arg \max_{S_k} P(S_k / Y_1, Y_2, \dots, Y_N) = \arg \max_{S_k} P(S_k / \bar{Y}_1, \bar{Y}_2, \dots, \bar{Y}_K)$$

Two recursions algorithm (BCJR algorithm)

$$P(S_k / Y_1, Y_2, \dots, Y_N) \propto P(S_k / \bar{Y}_1^k) \times P(\bar{Y}_{k+1}^K / S_k)$$

$$P(S_k / \bar{Y}_1^k) \propto P(\bar{Y}_k / S_k) \cdot \sum_{S_{k-1}} P(S_k / S_{k-1}) \cdot P(S_{k-1} / \bar{Y}_1^{k-1})$$

$$P(\bar{Y}_k / S_k) = P(\bar{Y}_k / \bar{U}_k)$$

Local measurements;
Channel model

Soft Decoding of VLCs

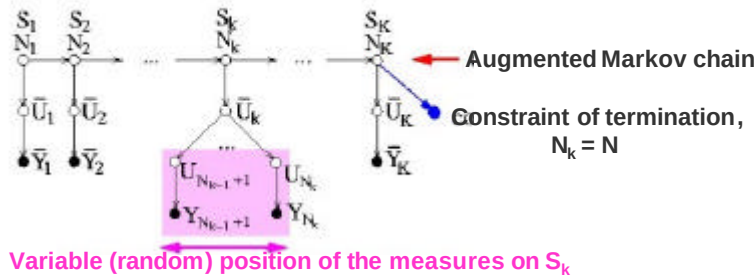
- **Symbol clock model for VLCs:**

- With VLCs, one has to locate S_k in the bitstream to compute the local measurements:

$$P(\bar{Y}_k | S_k, N_k) = P(\bar{Y}_{N_{k-1}+1}^{N_k} | \bar{U}_{N_{k-1}+1}^{N_k})$$

- Joint problem of segmentation and estimation

$$N_k = N_{k-1} + \mathcal{L}(S_k)$$



Soft Decoding of VLCs

- 1- Bit clock : estimate $(X_n, K_n) = (v, k)$ assuming independant symbols, exploiting the inner codeword correlation

⇒ State space of reduced dimension

⇒ But sub-optimal (inter-symbol correlation not exploited)

- 2- To transform the marginals on $(X_n, K_n) = (v, k)$ in marginals on $(S_k, N_k) \Rightarrow$ Clock conversion on soft information, given the recursion $N_k = N_{k-1} + \text{length}(S_k)$

- 3- To incorporate the inter-symbol correlation

$$P(\bar{Y}_k | S_k, N_k) = P(\bar{Y}_{N_{k-1}+1}^{N_k} | \bar{U}_{N_{k-1}+1}^{N_k})$$

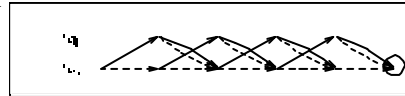
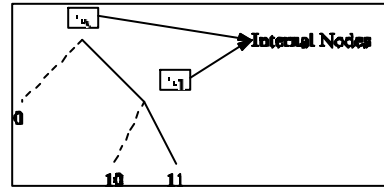
$$P(S_k | Y) \propto P(S_k | \bar{Y}_1^k) \cdot P(\bar{Y}_{k+1}^K | S_k)$$

$$P(S_k | \bar{Y}_1^k) \propto P(\bar{Y}_k | S_k) \cdot \sum_{S_{k-1}} P(S_k | S_{k-1}) \cdot P(S_{k-1} | \bar{Y}_1^{k-1})$$

This separate treatment of VLC coder and of Markov source is optimal

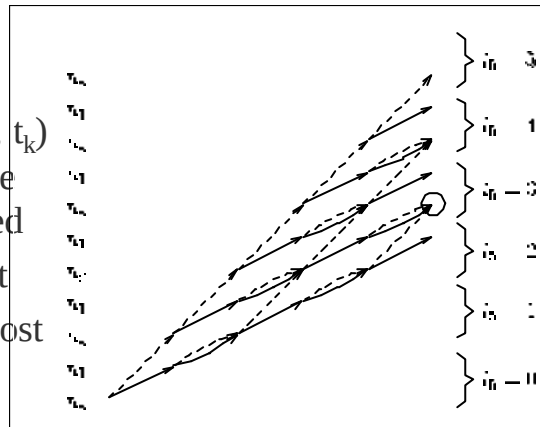
Trellis and complexity issues

- Bit-level trellis [Balakirsky 97]
 - The states are the internal nodes n_k
- Termination constraint
 - The trellis terminates in the root node
- Computing cost
 - $O(|A| \times L)$
- VLC tables can be simplified
 - [Kieffer et al 05]



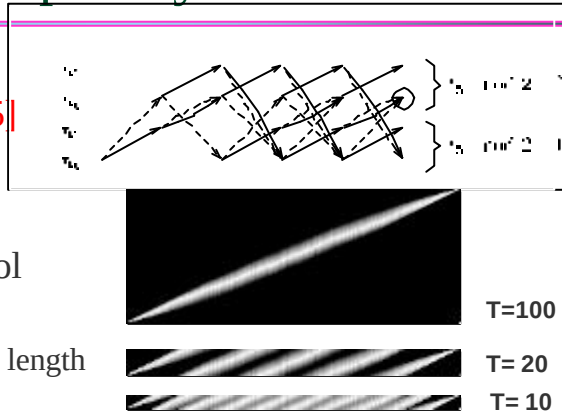
Trellis and complexity issues

- Bit/symbol trellis [Park&Miller 98, Hagenauer00]
- States are defined as (n_k, t_k) so that the memory of the symbol clock is preserved
- Symbol length constraint
- Prohibitive computing cost
 - $O(|A| \times L^2)$
 - Sub-optimal decoding



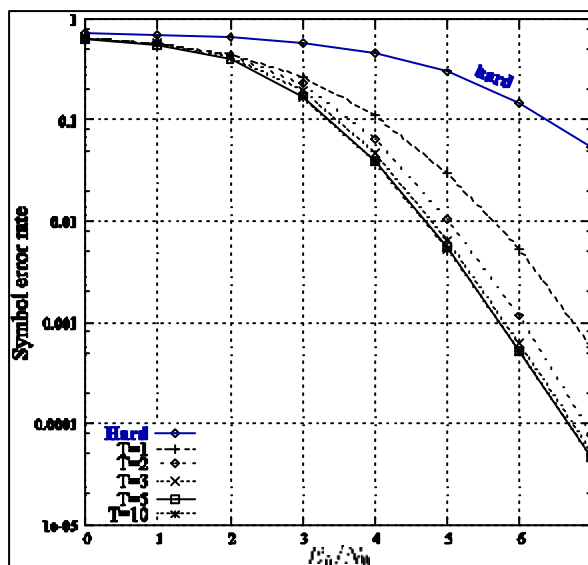
Trellis and complexity issues

- Aggregated trellis
[Jégou&Guillemot 05]
- States are defined as
 $(n_k, t_k \bmod T)$
- Decoding with Symbol length constraint
 - Partial information on length is preserved
- Computing cost
 - $O(|A| \times T \times L)$
 - Close to optimal decoding

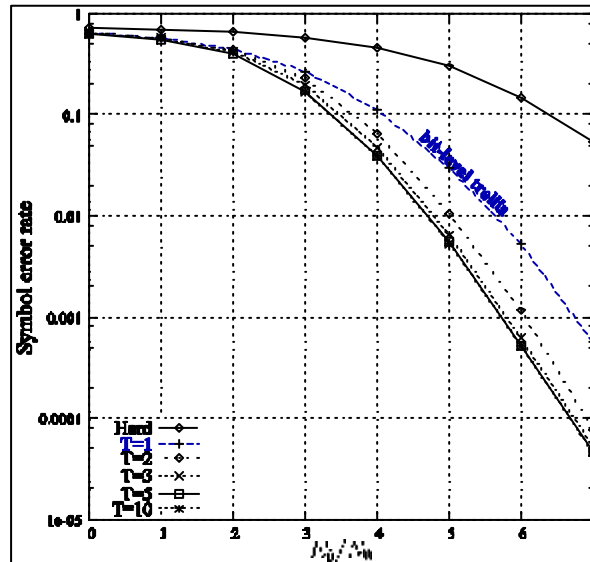


- $T=1$, bit-level trellis
- $T=L(S)$, bit/symbol trellis
- $1 < T < L(S)$, trade-off complexity / estimation reliability

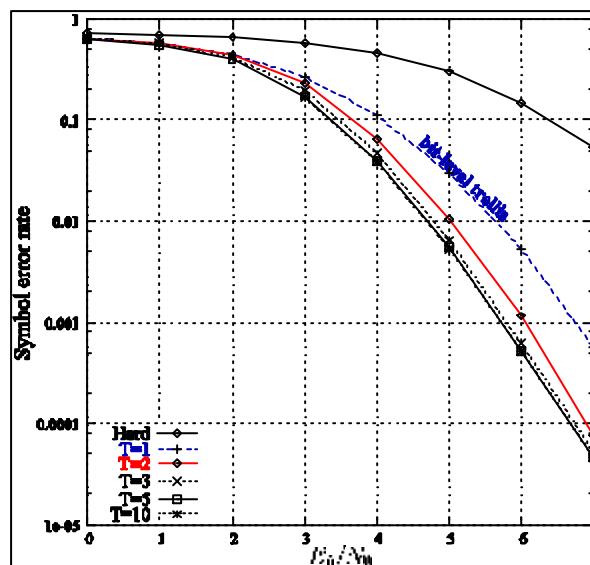
Performance of aggregated state model



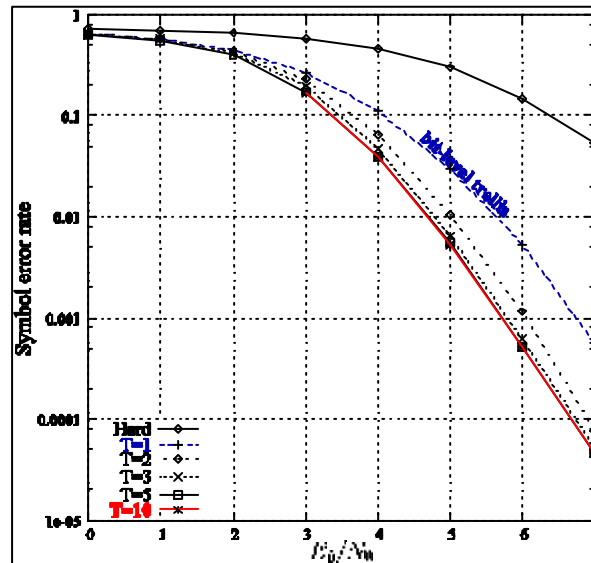
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Performance of aggregated state model

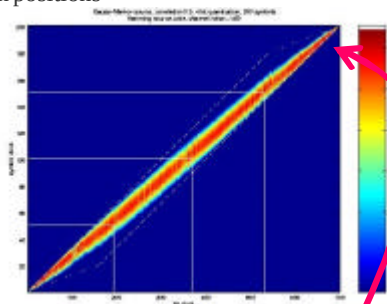
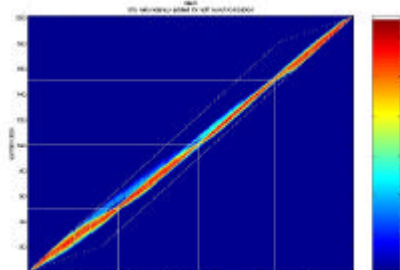
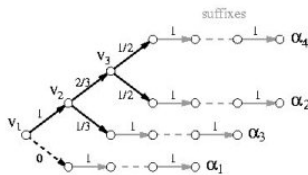


Performance of aggregated state model



Soft Synchronisation

- To add redundancy specifically dedicated to re-synchronization of bitstream and sequence of symbols
- A-priori known suffix to some symbols at known positions

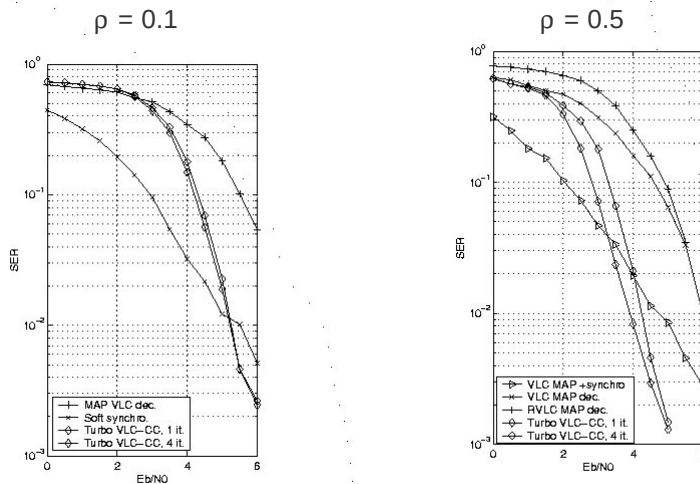


A posteriori Law
 $p(k, n \mid \text{measures})$

Termination constraint

RVLC useless except for a simple decoding

Performance of Soft Synchronisation

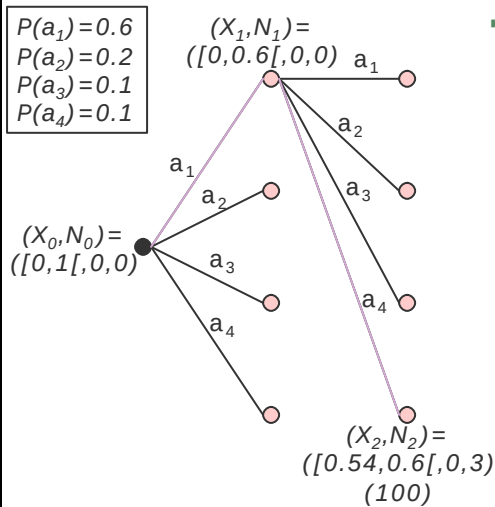


Arithmetic coding

- Arithmetic codes in new standards (H264, JPEG2000)
- Robust arithmetic coding/decoding
 - Augmentation of redundancy
 - Forbidden symbol for error detection & pruning
[Boyd & al. 97, Sodagar & al. 00, Pettijohn & al. 01...]
 - Forbidden error detection coupled with ARQ
[Elmasry 99, Chou & Ramchandran 00]
 - coupled with an ECC for error detection [Kozintsev & al. 98, ...]
 - Sequential decoding on decoding tree [Pettijohn & al. 01]

Soft Decoding of Arithmetic Codes

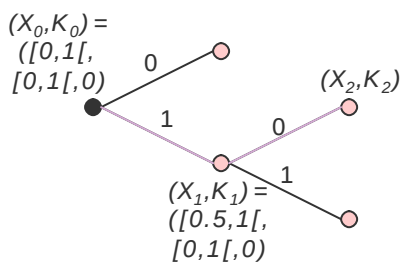
Arithmetic coder : M-ary tree



- Stochastic automaton driven by a symbol clock
 - State of the coding process $X_k = (\Pi_k, h_k[\cdot, nsc_k])$
 - Transitions triggered by a symbol
 - # of bits emitted per transition is random: Model augmented by a counter
 - Transition probabilities follow $P(S_k | S_{k-1})$

Soft Decoding of Arithmetic Codes

Arithmetic decoder : Binary tree

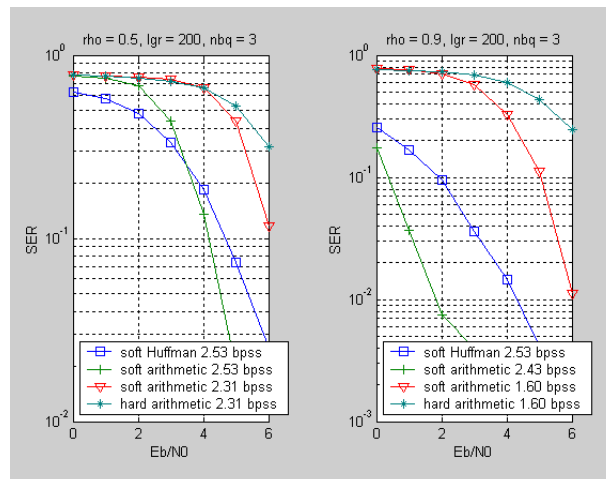


- Stochastic automaton driven by a BIT clock
 - State of the coding process $X_n = ([l_n, h_n[, [L_{Kn}, H_{Kn}[$
 - Transitions triggered by a BIT
 - # of symbols produced per transition is random: Model augmented by a counter
 - Transition probabilities follow $P(S_{K(n)+1} \dots S_{Kn} \mid S_1^{K(n-1)})$

Soft Decoding of Arithmetic Codes

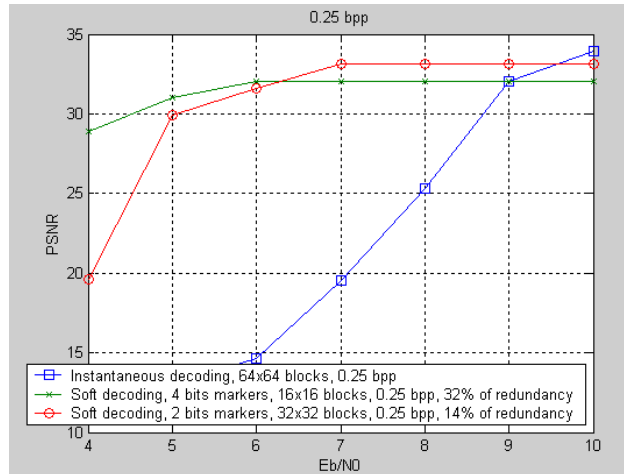
- Soft synchronization
 - Addition of bit patterns : a-priori info.
 - Increase the likelihood of synch. paths
 - Favor the pruning of some branches
 - Redundancy adjusted by size & freq.
- Exponential complexity (tree)
 - Adapted pruning strategy relying on
 - Likelihood of the paths
 - Available computing resources

Results on theoretical sources



Application to JPEG-2000 (EBCOT)

- Headers protected by a convolutional code with rate 1/3
- Possibility of markers addition up to *every stripe*
- Use of standard error detection markers as synchronization markers
- AWGN channel with BPSK modulation
- Results averaged over 100 realizations
- Lena at 0.25 bpp



Results with JPEG-2000 (0.25 bpp; $E_b/N_0=5$ dB)

Without errors
PSNR = 33.98 dB

With error resilience options
Without soft decoding
PSNR = 16.43 dB, 0.92 s.

With error resilience options
With soft decoding $W=10$
PSNR = 25.15 dB, 11.16 s.

$W=20$,
PSNR = 31.91 dB
23.8 s.



Application to CABAC in H.264

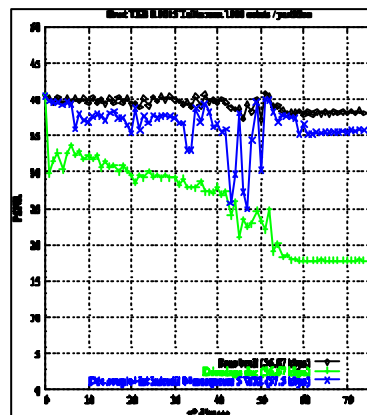
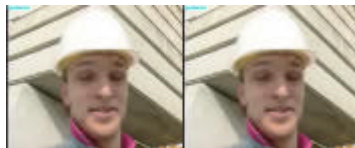
- APP computed for bits in one NAL unit
- Pruning technique
 - Only W paths with maximum probabilities are stored
 - Complexity controlled with W (W=1=> complexity of classical CABAC decoder)
- Termination constraint
 - The estimated sequence is the best path with the right number of transmitted bits and symbols
- With forbidden interval (FI), set of 2D tables storing the range of the LPS accounting for the width of the FI
- The LPS probability space (64 states) remains the same

Performance with H.264/CABAC

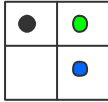
- Robust CABAC decoder: MAP estimation algorithm
- CABAC options: error detection & correction mechanism

Foreman QCIF@15fps (débit ~56 kbps), BER: 0.0015

- Classical decoding MAP decoding
(int 0 W 32 mark 5)

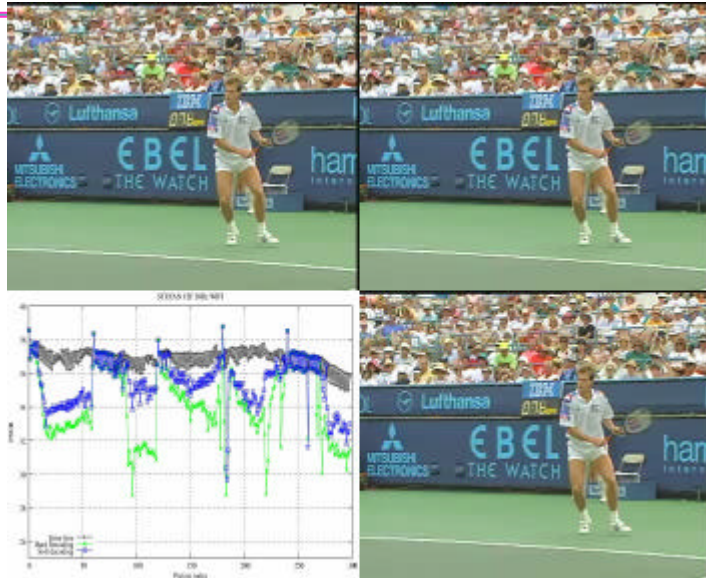


Performance with H.264/CABAC



Pattern WIFI
(802.11b)

- Without errors
- Classical decoding
- New mechanism

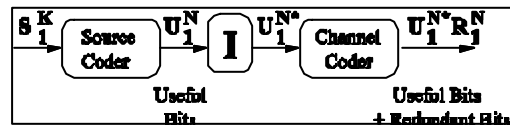


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Source-channel Turbo coding/decoding

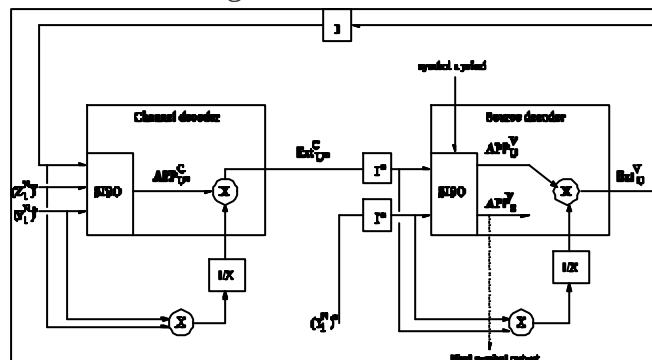
- Serial Turbo coding structure



- Direct connection of the 2 HMMs would result in a complex dependency (Bayesian) network with short cycles
=> Not amenable to fast estimation

Source-channel Turbo coding/decoding

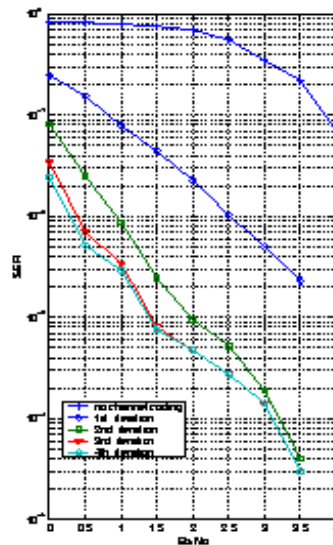
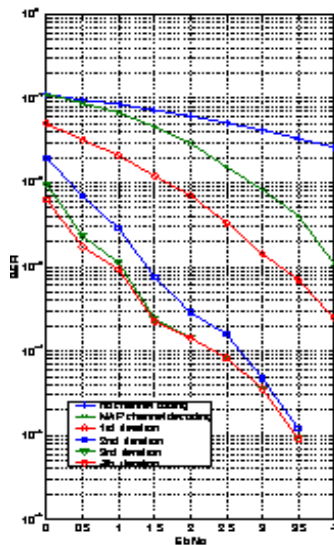
- Serial Turbo decoding structure



- Extrinsic information: modification induced by a new measure, here all but the local one on the APP of a soft-decoder output

$$Ext_{U_n}^V(Y = y | Y_n = y_n) = \frac{\mathbb{P}(U_n | Y = y)}{\mathbb{P}(U_n | Y_n = y_n) \underbrace{Ext_{U_n}^C(Y = y | Y_n = y_n)}_{\text{Initialized to 1}}}$$

Performance illustration of turbo decoding



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New source codes: Motivation

- To attain the capacity of a BSC channel the binary input of the channel should be uniform
- However
 - The output of VLCs is usually not uniform
 - Systematic error correcting codes => the uniformity condition is not satisfied
- Some authors proposed to use non-systematic turbo-codes [Zhy & Alajaji 02,04]
- The problem can be addressed from a source coding perspective

Variable Length Re-writing Systems

- Based on production rules (grammar) of the forms
(symbol,bits)? bits

$$r : a \bar{l} \rightarrow \bar{b},$$

$$a \in \mathcal{A}, \bar{l} \in \{0,1\}^*, \bar{b} \in \{0,1\}^+$$

- Extend the usual class of Variable Length Codes
- Small number of states for the encoding/decoding automata (for a given a priori distribution)
- Simple coding/decoding (table look-up)
- Improved joint performance for the joint source/channel decoding with these codes

Variable Length Re-writing Systems

- Examples of codes and tree representations:

$$r_{1,1} : a_1 \rightarrow 0$$

$$r_{2,1} : a_2 \rightarrow 10$$

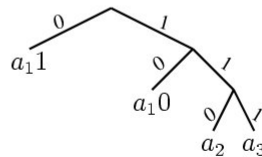
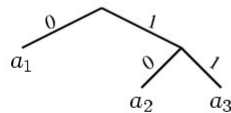
$$r_{3,1} : a_3 \rightarrow 11$$

$$r_{1,1} : a_1 1 \rightarrow 0$$

$$r_{1,2} : a_1 0 \rightarrow 10$$

$$r_{2,1} : a_2 \rightarrow 110$$

$$r_{3,1} : a_3 \rightarrow 111$$



- Decoder point of view: leaves = a symbol + some bits that are still to be decoded

Variable Length Re-writing Systems

Notation: $a \subset b$ iff a is a prefix of b

- The set $\bigcup_{i=1}^{|\mathcal{A}|} \bigcup_{j=1}^{|\mathcal{R}_i|} \{\bar{b}_{i,j}\}$ forms a prefix code
- $\forall i, \bigcup_{j=1}^{|\mathcal{R}_i|} \{\bar{b}_{i,j}\}$ is the set $\{\varepsilon\}$ or forms a full prefix code
- $\forall i \forall i' / i' \neq i, \forall j, j', \bar{b}_{i,j} = \bar{b}_{i',j'} \text{ or } \bar{b}_{i,j} \not\subset \bar{b}_{i',j'}$

These conditions are required to ensure that

- The code is uniquely decodable
- The code will be uniquely encodable with the proposed backward encoding procedure
- The bits output by the previous production rule suffice to select the next production rule

Encoding process

- Encoding is processed backward
- Initialization requires the encoding bit 1

$r_{1,1} : a_1 1 \rightarrow 0$	$r_{1,1} : a_1 \ a_1 \ a_1 \ a_1 \ a_1 \ 1$
$r_{1,2} : a_1 0 \rightarrow 10$	
$r_{2,1} : a_2 \rightarrow 110$	
$r_{3,1} : a_3 \rightarrow 111$	

Encoding process

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Encoding process

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$r_{2,1} : a_2 \rightarrow 110$	$r_{1,1} : a_1 \ a_1 \ a_1 \ 1 \ 0$
$r_{3,1} : a_3 \rightarrow 111$	$r_{1,2} : a_1 \ a_1 \ 0 \ 0$

Encoding process

- Encoding is processed backward
- Initialization requires the encoding bit 1

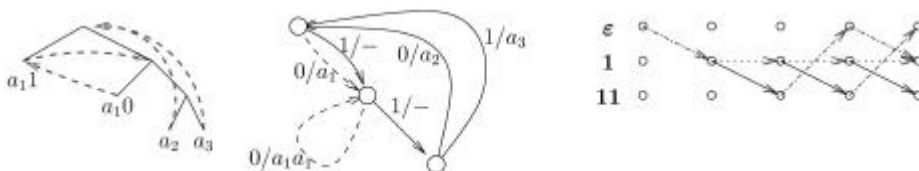
$$\begin{aligned} r_{1,1} : a_1 1 &\rightarrow 0 \\ r_{1,2} : a_1 0 &\rightarrow 10 \\ r_{2,1} : a_2 &\rightarrow 110 \\ r_{3,1} : a_3 &\rightarrow 111 \end{aligned}$$

$$\begin{aligned} r_{1,1} : a_1 \ a_1 \ a_1 \ a_1 \ a_1 \ 1 \\ r_{1,2} : a_1 \ a_1 \ a_1 \ a_1 \ 0 \\ r_{1,1} : a_1 \ a_1 \ a_1 \ 1 \ 0 \\ r_{1,2} : a_1 \ a_1 \ 0 \ 0 \\ r_{1,1} : a_1 \ 1 \ 0 \ 0 \end{aligned}$$

- Emitted sequence: 000

Variable Length Re-writing Systems

- Encoding and decoding can be implemented with sequential automata
- Definition of internal states
 - Bit sequence of variable length
 - Encoder: the smallest set of bit sequences that suffice to choose any rule
 - Decoder: a set of bits that do not generate a rule
- Example: tree, decoding automaton and trellis



Compression efficiency

- Compression efficiency: Expected number of bits produced by a production rule

$$\sum_{r_{i,j}} \delta_{i,j} \mathbb{P}(R_t = r_{i,j})$$

$$\delta_{i,j} = L(\bar{b}_{i,j}) - L(\bar{l}_{i,j})$$

- $(R_{L(S)}, \dots, R_t, R_{t+1})$ forms a Markov chain of transition probabilities

$$\mathbb{P}(R_t = r_{i,j} | R_{t+1} = r_{i',j'}) = \begin{cases} \mathbb{P}(a_i) & \text{if } \bar{l}_{i,j} \subset \bar{b}_{i',j'} \\ 0 & \text{otherwise.} \end{cases}$$

Compression efficiency

Assuming that $(R_t)_t$ is ergodic, $\lambda = \mathbb{P}(R_t)$ is obtained from $\pi = \mathbb{P}(R_t | R_{t+1})$ as the solution of the matricial equation

$$\lambda = \pi \times \lambda.$$

$\Rightarrow$ it leads to $E(L(\bar{B}_{i,j}) - L(\bar{L}_{i,j}))$, i.e. to the asymptotic compression efficiency

Example:

VLRS $\{a_1 1 \rightarrow 0, a_1 0 \rightarrow 10, a_2 \rightarrow 110, a_3 \rightarrow 111\}$,

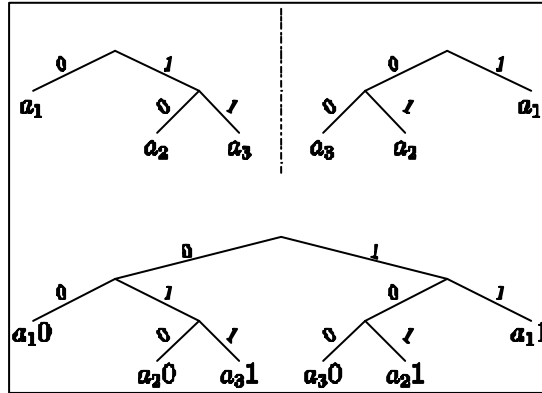
$\mathbb{P}(a_i) = \{0.7, 0.2, 0.1\} \Rightarrow$ expected length = 1.188 bits.

Huffman codes $\Rightarrow$ 1.3 bits.

In average, a_1 is coded with less than 0.5 bits.

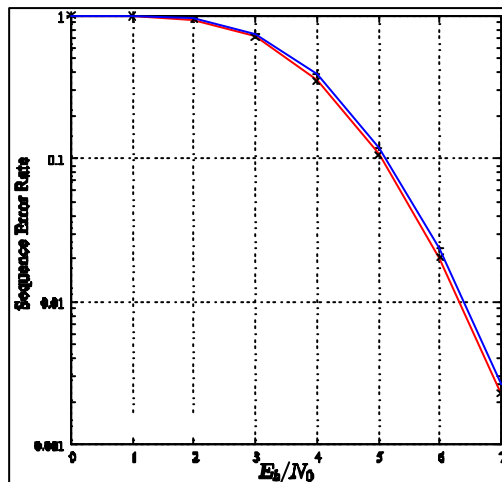
Mirrored construction

- To design a code leading to a uniform distribution of bits
- Construction from a VLC (same EDL)



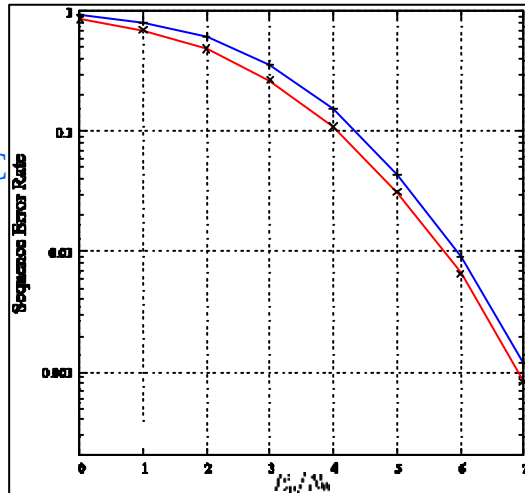
Soft decoding performance

- Code {00, 11, 010, 101, 0110}
- $P(0)=0.6$
- Mirror VLRS vs RVLC
- Balakirsky-like state model
- Viterbi decoding



Soft decoding performance

- Code {00, 11, 010, 101, 0110}
- $P(0)=0.917$
- Mirror VLRS vs RVLC
- Balakirsky-like state model
- Viterbi decoding



Outline

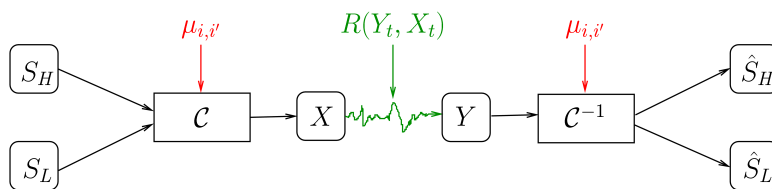
- Applications and problems addressed
- Soft decoding of VLCs
- Source-channel turbo coding/decoding
- Source codes with properties for source-channel coding
- Source codes with inherent resilience
- Conclusion

Multiplexed Codes, principle (1)

- Most of existing JSCC solutions introduce some redundancy in the encoded bitstream
- A new family of codes:
 - no redundancy
 - reduction of the impact of channel noise
 - assume the existence of two sources (S_H and S_L)
 - guarantee no error propagation for S_H (synchronous)
 - ® Inherent Unequal Error Protection (UEP)
- Design Principle :
 - redundant FLCs designed for S_H
 - redundancy used to describe S_L

1

Transmission scheme: 2 sources

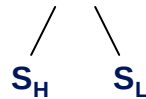


- S_H : First-order Markov source, of high priority
 - Takes its values on a finite alphabet $A=\{a_1...a_p,...\}$
 - Transitions probabilities $\mu_{i,i'} = P(S_t=a_i / S_{t-1}=a_{i'})$
- S_L : uniform memoryless binary source of low priority
- Channel: Discrete Memoryless Channel
 - $R(Y_t, X_t) = P(Y_t \text{ received} / X_t \text{ emitted})$

Multiplexed Codes, principle (2)

- Alphabet A of S_H : $a_1 \dots a_j \dots a_W$, cardinality W
- Codewords of length c
 - ▷ set X of 2^c codewords
- X partitioned in W equivalence classes associated to symbols of S_H : $C_1, \dots, C_j, \dots, C_W$
Let $N_j = \text{card}(C_j)$
- Bijection:

X	$\mathbb{R}$	$A \times [0..N_j-1]$
cwd	$\mathbb{R}$	(a_j, q)



1

Stationary multiplexed codes

- Redundant FLC

- C_i : equivalence class of a_i

- q_i : index of a codeword within its equiv. class

- Inherits FLC properties

- Strict sense synchronization
- Random data access

a_i	μ_i	codeword	index q_i	
a_1	0.4	000	0	} C_1
		001	1	
		010	2	
a_2	0.2	011	0	} C_2
		100	1	
a_3	0.2	101	0	} C_3
a_4	0.1	110	0	} C_4
a_5	0.1	111	0	} C_5

Compression efficiency

- For each codeword:
 - c bits used
 - index variable Q_i : $\log_2(N_i)$ bits for S_L
 - ▷ a symbol of S_H is coded with $c - \log_2(N_i)$ bits
- Mean description length (mdl) of S_H :
 - $\sum m_i \log_2(N_i / 2^c)$
- N_i are chosen to reach the entropy, i.e closest to
 - $N_i \approx m_i 2^c$

1

Multiplexing capacity

- Number L of sequences representing a given sequence s_H of length K :
 - $L = \sum_{t=1..K} N_t$
- Theoretical multiplexing capacity: $\log_2(L)$ bits
- How to use this capacity?
 - Ⓡ conversion of S_L into a sequence of states q_t ($0..N_i-1$)
- Assumption:
 - S_L already encoded as a bitstream b

1

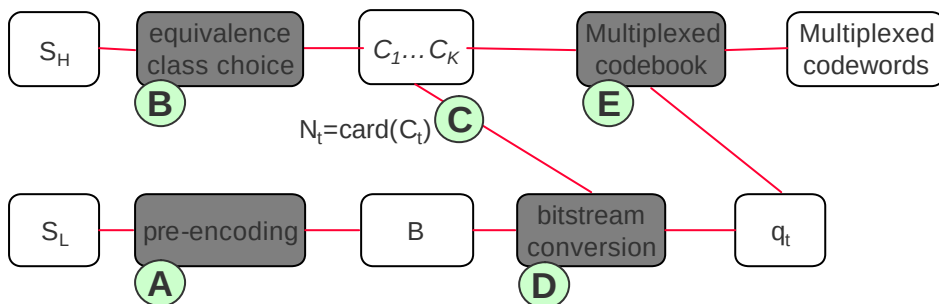
Multiplexing capacity

s_H	a_1	a_4	a_5	a_2	a_3	a_3	a_1	a_2
	000	110	111	011	101	101	000	011
C_i	001			100			001	100
	010						010	
$ C_i $	3	1	1	2	1	1	3	2

- Number of sequences representing s_H : $3*2*3*2=36$
- Multiplexing capacity: $\log_2(36)=5.17$ bits
- s_H is finally coded with $3*8-|5.17|=19$ bits

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Encoding scheme (1)



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Encoding scheme (2)

- (A) S_L is pre-encoded as a bitstream b (compressed)
- (B) "t, the realization s_t of S_H indexes an equivalence class
- (C) Processing N_t provides the cardinality of the index variable Q_t
- (D) The bitstream b is converted into a sequence of N_t -valued variables
- (E) Couples (a_1, q_t) provides entries in the multiplexed table

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Step (D): conversion of b (1)

□ First approach:

- b ° binary representation of an integer g
- g : realization of a L -valued variable, indexing codewords of the whole sequence for a realization
- g is transformed into a sequence of state (q_t) using Euclidean divisions by N_t

$$g = q_1 + N_1 (q_2 + N_2 (\dots + N_t (q_{t+1} + \dots N_{K-1} q_K) \dots) \dots))$$

□ Main Advantage & Drawback

- Optimality in terms of compression (depending on c)
- but dealing with long integers

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Step D : conversion of b (2)

■ 2nd approach:

- no calculation of a global state
- states (q_i) are processed using transformations with low computing cost
- Not optimal in compression: overhead < 1% to entropy

■ Binary multiplexed codes:

- Bits of b directly index codewords
- $N_i = 2^i$

■ MUX-codes constructed from VLC codetrees

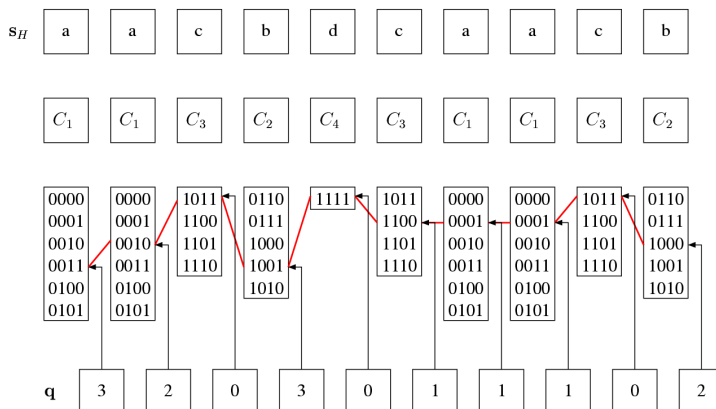
- $SER(BER) = BER \cdot mdl + O(BER)$ on a BSC

1

Step D: Conversion of S_L : problem statement

- Several codeword sequences are available

® bijection with a sequence of N_t -valued states (q_i)



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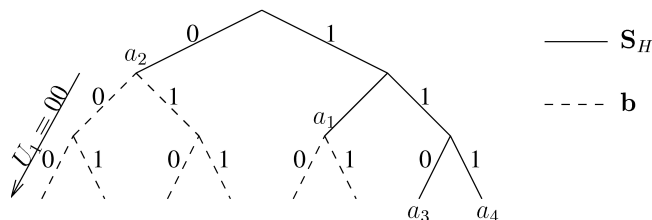
Binary multiplexed codes (1)

- **Is it possible to avoid step D?**: Yes, Binary Multiplexed Codes
- **Definition :**
 - a multiplexed code is binary iff " i ", N_i is a power of 2
- **Consequences :**
 - codewords in a given equivalence class can be directly indexed a segment U_i of $\log(N_i)$ bits
 - ↓ no bitstream conversion is required
- **Compression efficiency of Huffman codes**

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Construction from a VLC codetree

- The VLC codetree is completed to reach the longest codeword length
- A codeword prefix is used for S_H
- A codeword suffix "store" some bits U_t of B



- **SER performance on a BSC :**

$$SER(BER) = BER \cdot m d / + O(BER)$$

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Binary multiplexed codes (2)

class $\mathcal{C}_i$	cwd $c_{i,q}$	a_i	N_i	$\mathbb{P}(a_i)$	index q	or U_i
$\mathcal{C}_1$	000	a_1	2	0.30	0	0
	001				1	1
$\mathcal{C}_2$	010	a_2	4	0.43	0	00
	011				1	01
	100				2	10
	101				3	11
$\mathcal{C}_3$	110	a_3	1	0.25	0	$\emptyset$
$\mathcal{C}_4$	111	a_4	1	0.02	0	$\emptyset$

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Encoding example

$$\begin{aligned} \mathbf{s}_H &= \mathbf{a}_1 \mathbf{a}_2 \mathbf{a}_2 \mathbf{a}_3 \mathbf{a}_2 \mathbf{a}_1 \mathbf{a}_2 \mathbf{a}_4 \\ \mathbf{b} &= 0101010101 \end{aligned}$$

$\mathbf{p}$

$$\log(N)_{t=1..8} = 1 \ 2 \ 2 \ 0 \ 2 \ 1 \ 2 \ 0$$

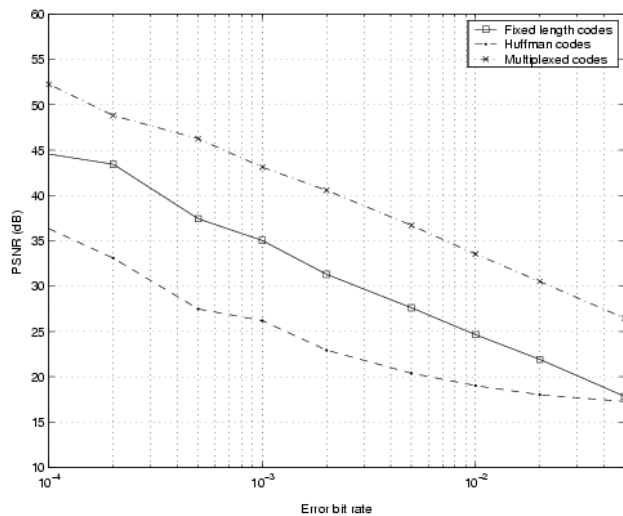
$$(U)_{t=1..8} = (0, 10, 10, \emptyset, 10, 1, 01, \emptyset)$$

Using the multiplexed codebook, the generated bitstream is

000 100 100 110 100 001 011 111

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Multiplexed Codes



FLC



Huf



Mul



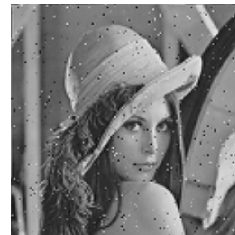
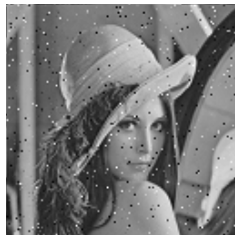
Simulation results (BSC)

FLC (uncompressed)

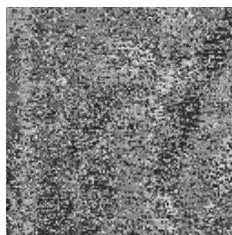
Huffman

MUX codes

BER=
0.01



BER=
0.05



Conclusion on Stationary MUX-codes

- De-synchronization confined to the low priority source
- Source UEP
- Better SER and SNR results than FLCs for S_H
- Random data access for S_H
- Low complexity soft decoding for S_H

Is it possible to exploit higher-order source statistics?

First-order multiplexed codes

- Motivation: to exploit the conditional probabilities $m_{i,j}$
- Code construction
 - $\forall a_i \in A$, a multiplexed code is designed for the conditional pdf $m_{i,j}$
 - Equivalence classes indexed by $N_i^{j'}$
- For a stationary discrete first-order Markov source, the conditional entropy is reached iff

$$\forall (a_i, a_{i'}) \in A^2 \quad |N_i^{j'}| = P(a_i / a_{i'}) \quad |X|$$

Example

$\mu_{i,j}$	a_1	a_2	a_3	S_{t-1}	a_1	a_2	a_3
a_1	3	1	1	000			
	4	4	4	001			
a_2	1	1	1	010			
	8	2	4	011			
a_3	1	1	1	100			
	8	4	2	101			
				110			
				111			

$entropy_0 = 1.5$
 $entropy_1 = 1.28$
 $mdl = 1.28$

Index Assignment (IA)

- A symbol S_t is not correctly reconstructed in case of
 - Error in the codeword itself
 - Error in the previous reconstructed symbol
 => Error propagation may occur on contexts
- The IA conditioned by a given context has to be optimized by taking into account the other contexts
- Some codewords may represent the same symbol for all the previous conditioning values
 ? The set of codewords that have this property is called **kernel**

Kernel

- **Definition:** the kernel is the set of codewords $k =$

$$\{x \in \mathcal{X} / \exists a_i \in A, \forall a_i \in A, x \in C_i\}$$

- *In the example, $|k| = 3$*
- *It could be 4*

S_{t-1}	a_1	a_2	a_3
000	a_1	a_1	a_1
001	a_1	a_1	a_1
010			
011			
100			
101			
110			
111	a_3	a_3	a_3

Kernel

- **Definition:** the kernel is the set of codewords $k =$

$$\{x \in \mathcal{X} / \exists a_i \in A, \forall a_i \in A, x \in C_i\}$$

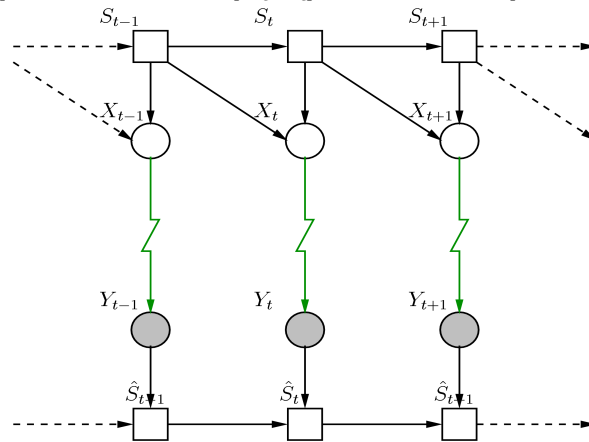
- *In the example, $|k| = 3$*
- *It could be 4*

S_{t-1}	a_1	a_2	a_3
000	a_1	a_1	a_1
001	a_1	a_1	a_1
010	a_2	a_2	a_2
011			
100			
101			
110			
111	a_3	a_3	a_3

The kernel offers natural hard synchronization properties

Error resilience analysis (S_H)

- The product model $(S_t, \hat{S}_t)$ is a Markov process



Error resilience analysis



- Transition probabilities of this model given by

$$P(\hat{S}_t = a_j, S_t = a_i | \hat{S}_{t-1} = a_{j'}, S_{t-1} = a_{i'}) = \frac{\mu_{i,i'}}{N_{i'}^{i', X_t, Y_t, \hat{S}_t}} \sum_{C_t^i, \hat{C}_t^j} R_i(Y_t, X_t)$$

- If this matrix positive, the asymptotic marginal probability $P(S_t = a_i, \hat{S}_t = a_j)$ can be computed from this matrix (Perron-Frobenius Theorem) (normalized eigenvectors associated to the eigenvalue 1)

$$SE_{K \rightarrow \infty} = 1 - \sum_{a_i \in A} P(\hat{S}_t = a_i, S_t = a_i)$$

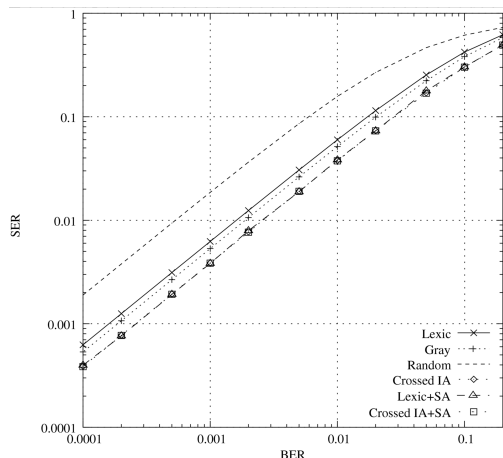
$$MSE_{K \rightarrow \infty} = \sum_{(a_i, a_j) \in A^2} P(\hat{S}_t = a_j, S_t = a_i) (a_j - a_i)^2$$

Cross-Index Assignment

- Objective:
optimizing the analytical expressions of SER and SNR
- Efficient optimization algorithms in the literature:
 - Binary Switching Algorithm (BSA) [Zeger & al. 90]
 - Simulated Annealing (SA), e.g. [Farvadin & al. 91]
- For the SER criteria, Simulated Annealing turns out to maximize the cardinal of the kernel

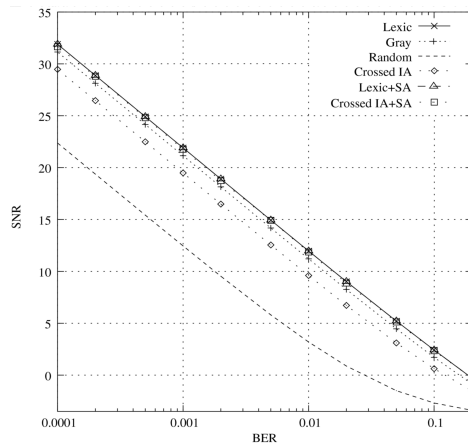
Symbol Error Rate on a BSC

- S_H : Gauss-Markov source
of correlation $\rho = 0.5$
- 8 quantization cells
- $H_1(S_H) = 2.29$
- Binary symmetrical
channel
- 1st-order MUX-code:
 - $c=5$
 - $mdl=2.35$



Signal to Noise Ratio on a BSC

- S_H : Gauss-Markov source of correlation $\rho=0.5$
- 8 quantization cells
- $H_1(S_H)=2.29$
- Binary symmetrical channel (BSC)
- 1st-order MUX-code:
 - $c=5$
 - $mdl=2.35$



Compression assessment in SVC

- Texture residue coded in CAVLC
 - High priority *NumTrail* (4 VLC tables depending on context)
 - Low priority *TrailingOnes* sign, *RunBefore*, *TotalRun*

	Tab 1	Tab 2	Tab 3	Tab 4
Entropy	2.1543	3.3809	4.6972	5.4692
Mdl cavlc	2.1620	3.4062	4.7749	6 (FLC)
Mdl huffman	2.1608	3.4062	4.7338	5.5040
Mdl multiplexed codes	2.1548	3.3871	4.6972	5.4693
Symbols distribution	62%	25%	10%	1%

- No loss in compression efficiency: error resilience for free!!

Conclusion

- Significant gains in SER/PSNR in exploiting existing VLC sub-optimality
 - High gains if soft measures available in the application layer
- Extra gain in keeping/inserting redundancy in the chain
 - A priori source information
 - Error correcting code
 - This has a cost in bandwidth
- Codes with inherent resilience
 - Almost no loss in compression efficiency
- All this makes “lite” mechanisms in lower layers meaningful to save bandwidth