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# HARQ Protocol for Burst Transmission over FSO Turbulence Channels

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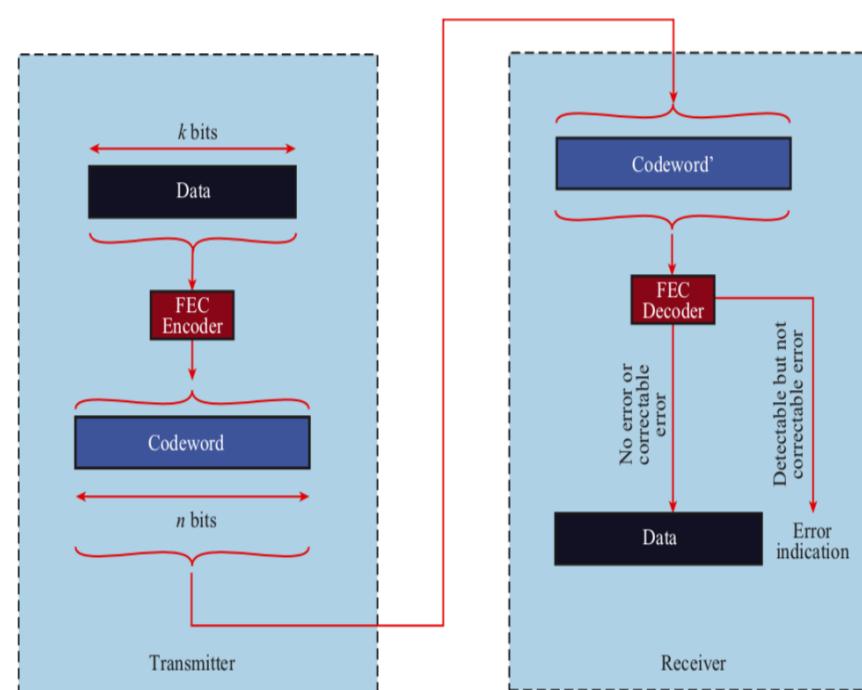
# Outline

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- Part I: Convolutional Codes
- Part 2: My study

# Part I: Review of FEC Code

- FEC is used to **correct** transmission errors over an unreliable and noisy communication channel **without asking for retransmission**
- Idea:
  - Transmitter encodes data by adding some **redundant bits**
  - Receiver, using redundant bits, can correct errors
- There are two main types of FEC
  - Block code: systematic code
  - Convolutional code: non-systematic code



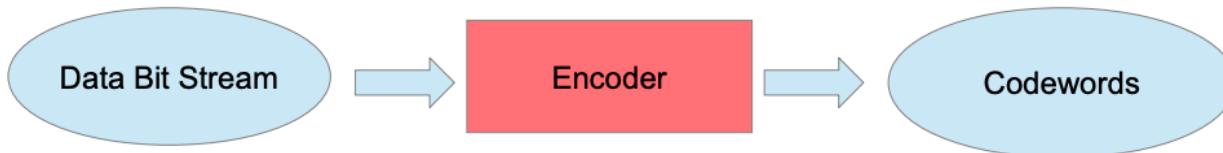
# Convolutional Codes (CC)

- For applications which require a **continuous stream** of bits (e.g. Digital video Broadcasting-Terrestrial), the use of **block codes** may not convenient.

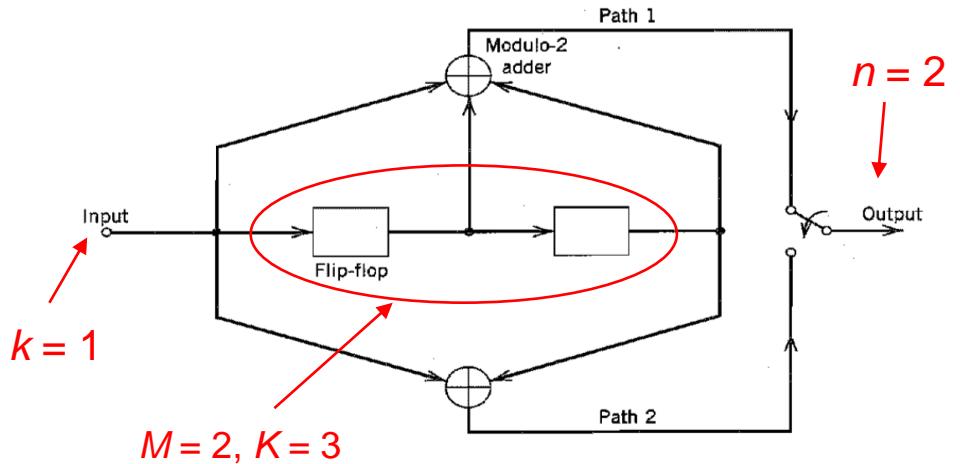


- The **convolutional codes**, that generate redundant bits continuously so that error checking and correcting are carried out continuously, are used for those applications.
- **Features:**
  - generates redundant bits by using *modulo-2 convolutions* (name of code)
  - has memory: output bits depend on not only the current input bits but also the previous bits
  - Non-systematic code: cannot distinguish the message bits and redundant bits

# Encoder of CC



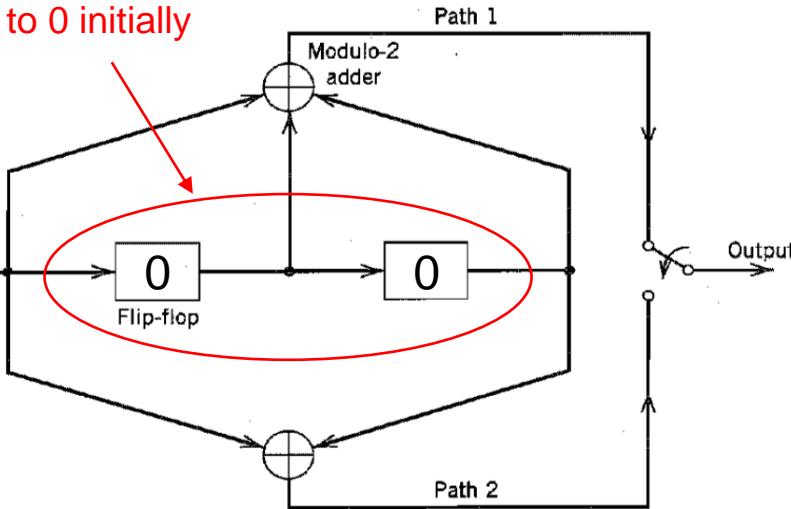
- A convolutional code ( $n, k, K$ )
  - $k$ : no. of message bits shifted into the encoder at a time ( $k = 1$  is usually used)
  - $n$ : no. of encoder output bits corresponding to the  $k$  message input bits
  - $K$ : constraint length; no. of shifts over which a single message bit can influence the encoder output ( $K = M + 1$ );  $M$ : shift registers
  - Coding rate:  $R_c = k/n$
- E.g. CC (2, 1, 3)



# Example of CC

- $(n, k, K) = (2, 1, 3)$

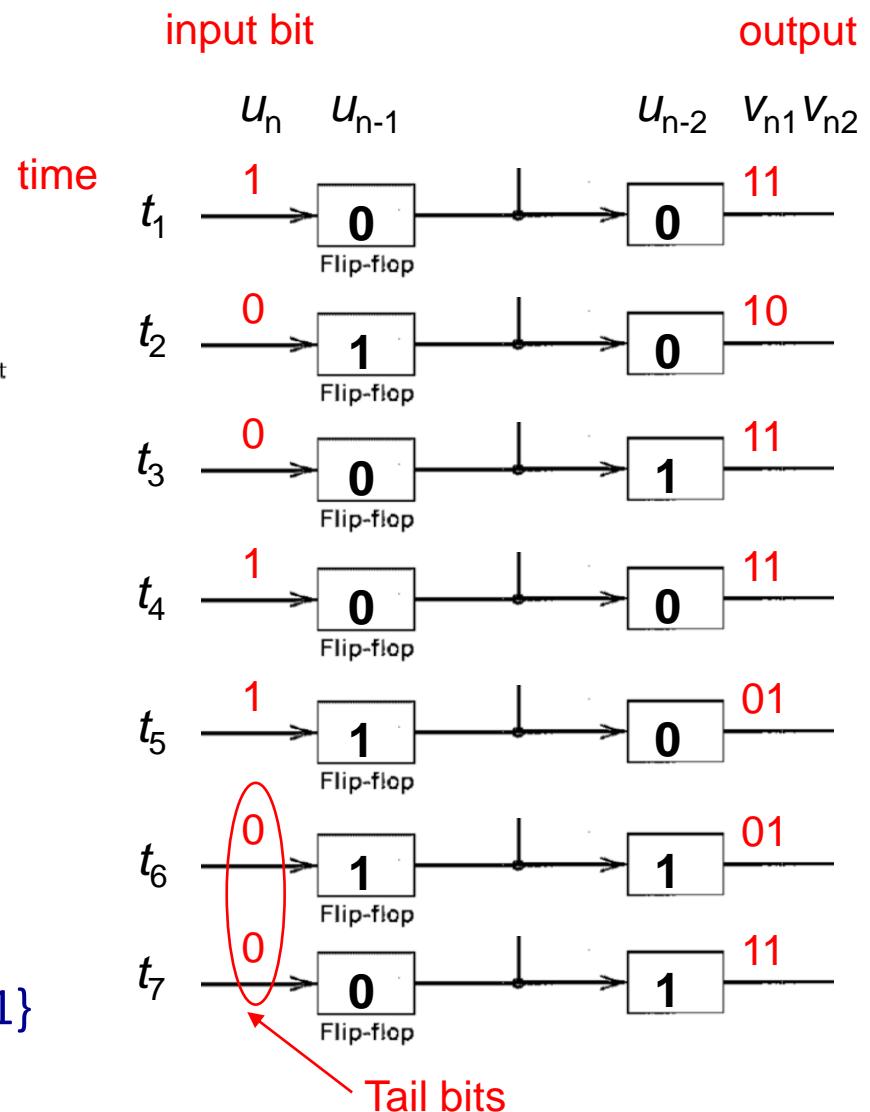
set to 0 initially



$$v_{n1} = u_n \oplus u_{n-1} \oplus u_{n-2}$$

$$v_{n2} = u_n \oplus u_{n-2}$$

- Input:  $m = 10011$
- Output:  $c = \{11, 10, 11, 11, 01, 01, 11\}$

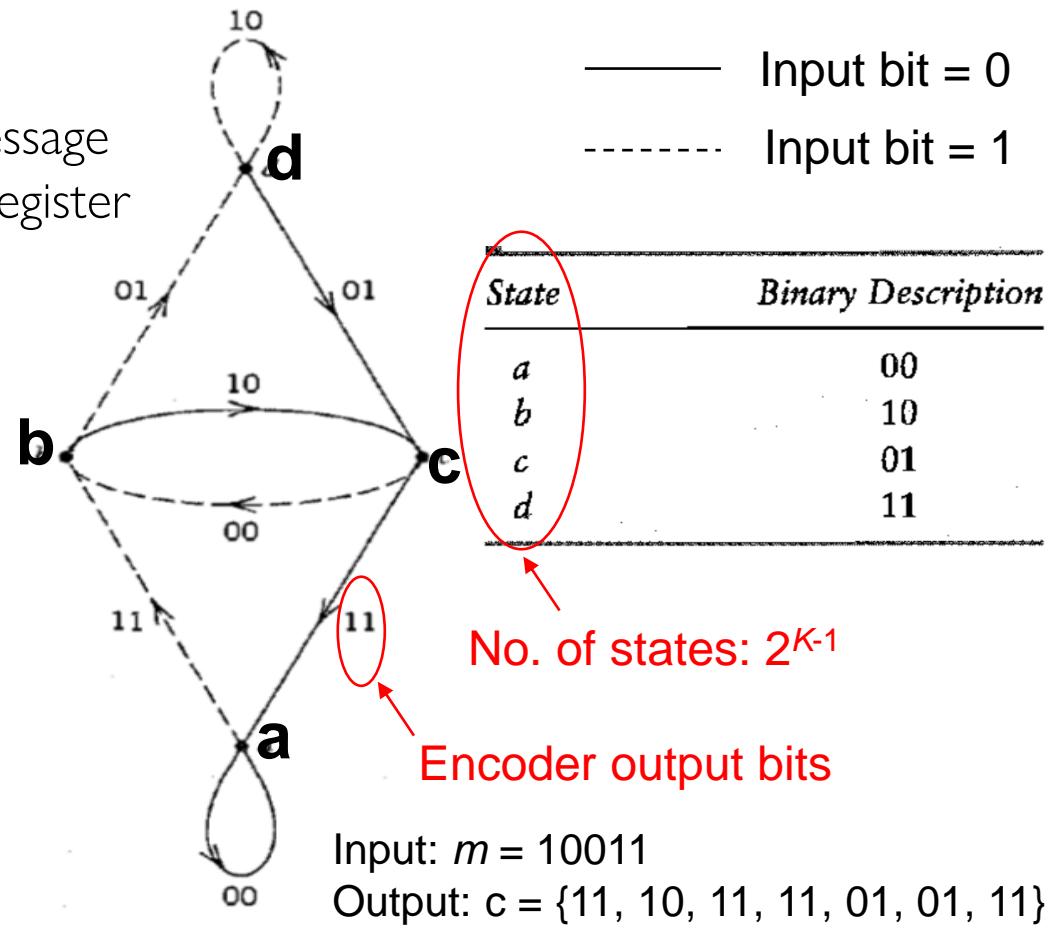
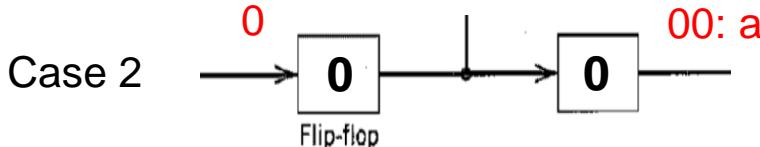
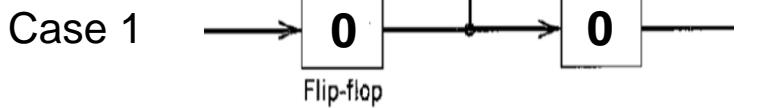
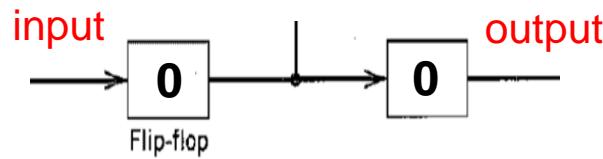


# Representation of CC (1)

- The structure properties of a convolutional encoder can be illustrated in graphical form such as (1) state diagram (2) trellis

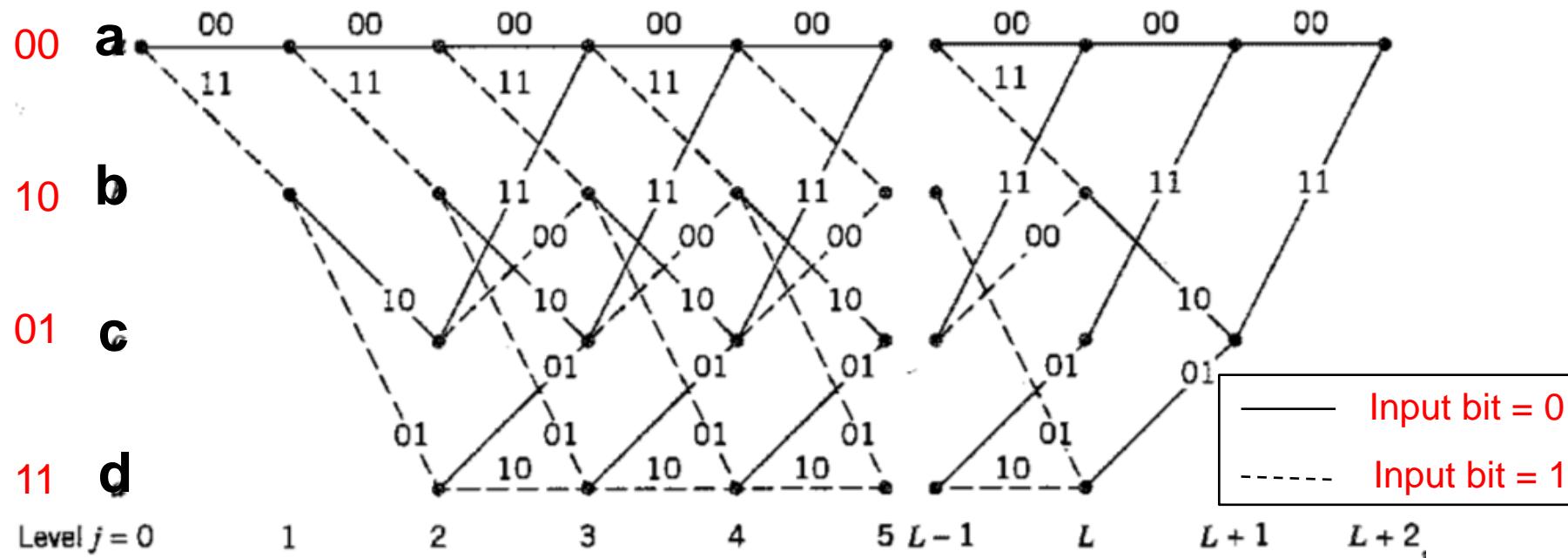
## (1) State diagram

- States are defined as  $(K-1)$  message bits stored in encoder's shift register
- E.g.: current state a: 00



# Representation of CC (2)

- (2) Trellis: extension of state diagram according to time



- The trellis contains  $(L + K)$  levels, where  $L$  is the length of incoming message
- Example:  $m = 10011 \rightarrow c = \{11, 10, 11, 11, 01, 01, 11\}$

# Decoder of CC

- How to get the correct message at destination?



- There are two kinds of algorithm to encode the CC codes
  - Maximum likelihood algorithm
  - Viterbi algorithm

# Maximum likelihood (ML) decoding (1)



- Given the received sequence  $r$ , the decoder is required to make an estimate  $\hat{m}$  of  $m$  (note:  $\hat{m} = m$  if and only if  $\hat{c} = c^{(m)}$ ).
- The decoding rule is the selection of the estimate  $\hat{c}$  so that the probability of decoding error ( $P_e$ ) is minimized.
- $P_e$  is minimized if the estimate  $\hat{c}$  is chosen to maximize the **likelihood function**,  $p(r | c)$

$$p(r|\hat{c}) = \max_{\text{over all } c} p(r|c)$$

where  $c$  is one of the possible transmitted sequences

# Maximum likelihood (ML) decoding (2)

- For a binary symmetric channel, both  $c$  and  $r$  represent binary sequences of length  $N$ , we have :  $p(r|c) = \prod_{i=1}^N p(r_i|c_i)$

where  $r_i$  and  $c_i$  are the  $i$ -th elements of  $r$  and  $c$

- The log-likelihood:  $p(r|c) = \sum_{i=1}^N p(r_i|c_i)$ ; where  $p(r_i|c_i) = \begin{cases} p, & \text{if } r_i \neq c_i \\ 1 - p, & \text{if } r_i = c_i \end{cases}$
- Suppose that  $r$  differs from  $c$  in exactly  $d$  positions;  $d$  is the *Hamming distance*, then we may re-write the log-likelihood,

$$\log p(r|c) = d \log p + (N - d) \log(1 - p) = d \log\left(\frac{p}{1-p}\right) + N \log(1 - p)$$

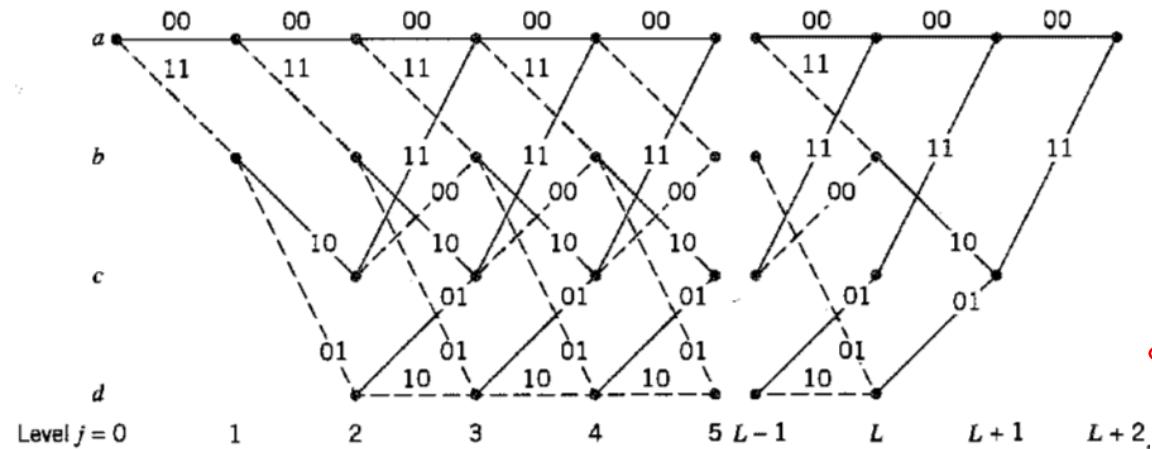
- In summary, the maximum-likelihood decoding rule for binary symmetric channel as follows,

Choose the estimate  $\hat{c}$  that minimizes the Hamming distance between the received sequence  $r$  and the transmitted sequence  $c$

# ML decoding: Example

- E.g.,  $m = 101 \rightarrow c = \{11 10 00 10 11\}; p = 0.1$ ;

received sequence  $r = \{11 11 00 10 11\}$ . Find  $\hat{m}$ ?



path	Code sequence	Hamming distance
00000	00 00 00 00 00	7
00100	00 00 11 10 11	6
01000	00 11 10 11 00	6
01100	00 11 01 01 11	5
10000	11 10 11 00 00	6
10100	11 10 00 10 11	1
11000	11 01 01 11 00	5
11100	11 01 10 01 11	4

- ML decoding is too complex to search all available paths (in case of very long input message bits)

- End-to-end calculation

**Viterbi algorithm performs ML by reducing its complexity**

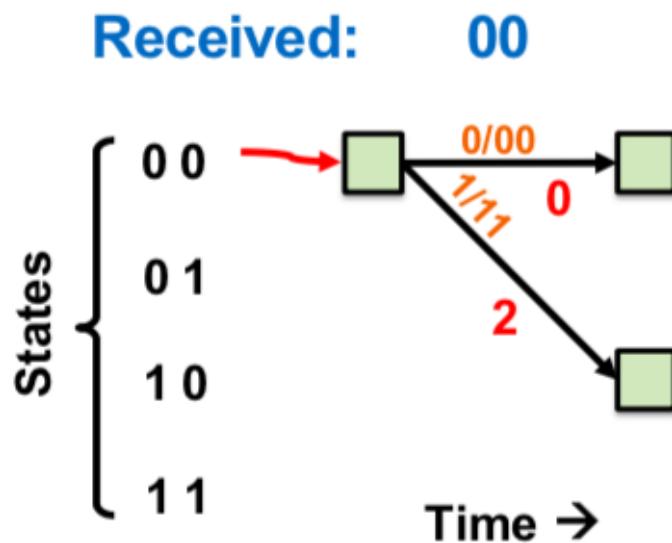
# Viterbi Algorithm

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- Viterbi reduces decoding complexity by removing the trellis paths that could not possibly be candidates for ML choice (early rejection)
- Origin of Viterbi Decoding
  - Andrew J. Viterbi, "Error Bounds for Convolutional Codes and an Asymptotically Optimum Decoding Algorithm," *IEEE Transactions on Information Theory*, Volume IT-13, pp. 260-269, April 1967.
  - Viterbi is a founder of Qualcomm.
- There are two kinds of Viterbi Decoding
  - Hard-decision Viterbi Algorithm
  - Soft-decision Viterbi Algorithm

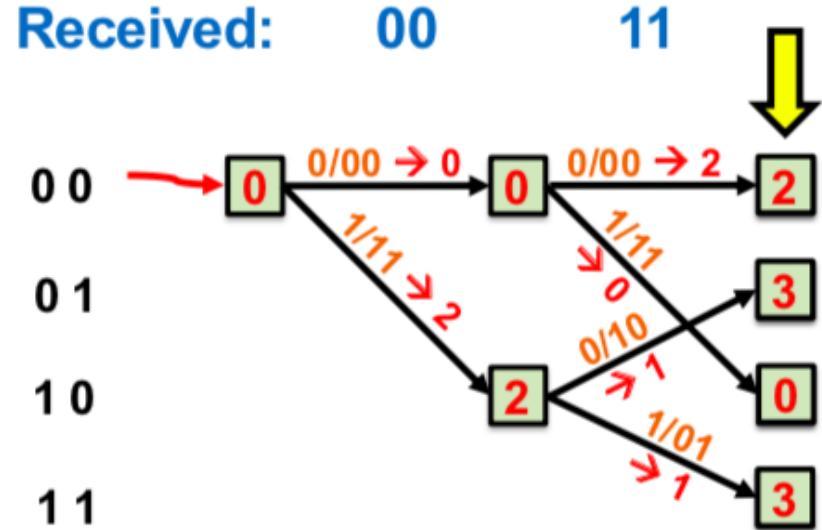
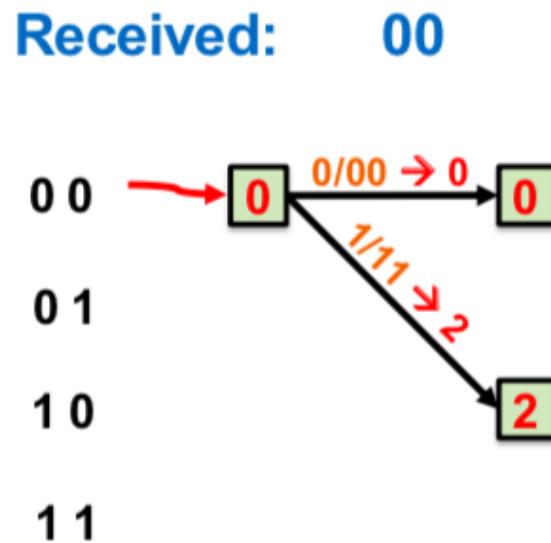
# Hard-decision: branch metric

- Branch metric = Hamming distance between received and transmitted bits
- Encoder is initially in state 00, receive bits: 00



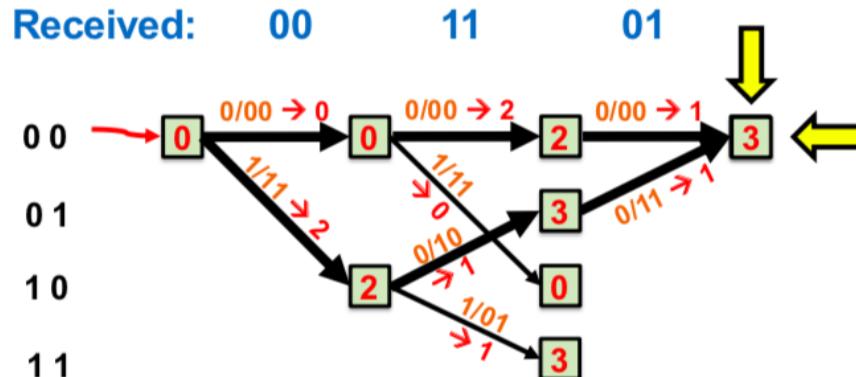
# Hard-decision: path metric

- Path metric = path metric of predecessor + branch metric
- Note: path metric for the left-most state of the trellis is 0

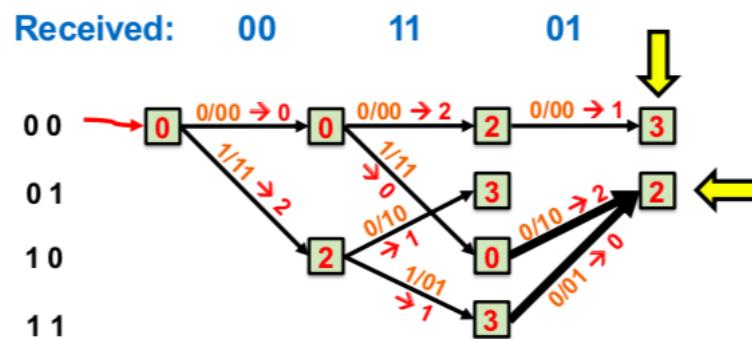


# Hard-decision: early rejection

- Problem: each state has two predecessors (or two branches enter a node)

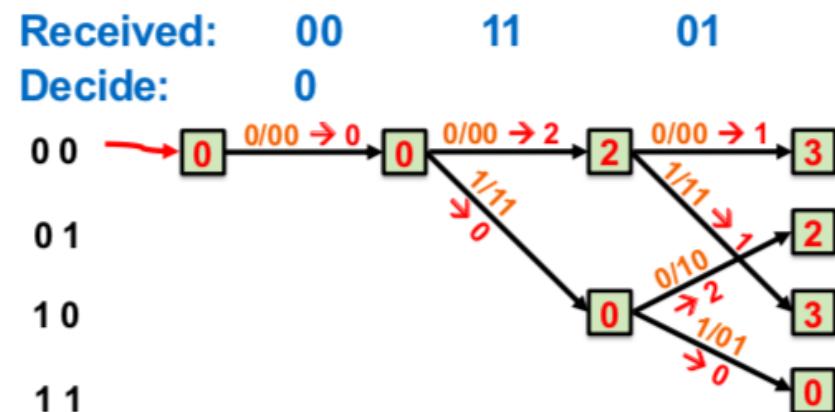
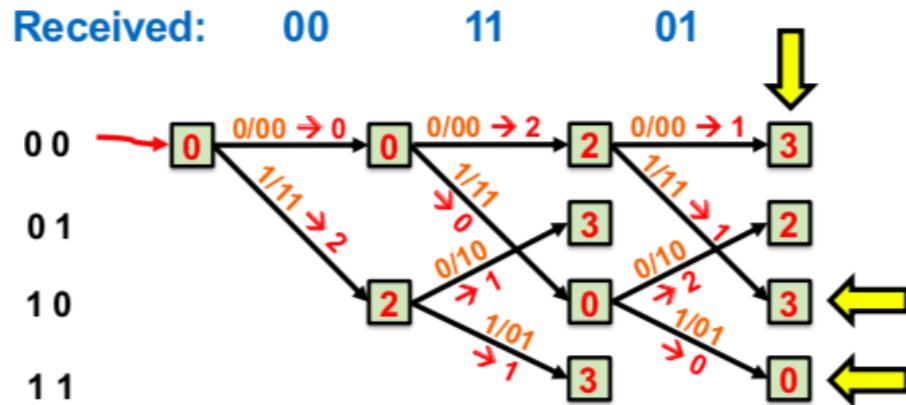


- The algorithm compares two path metrics corresponding to two predecessors. The path with lower metric is retained, and the other path is discarded



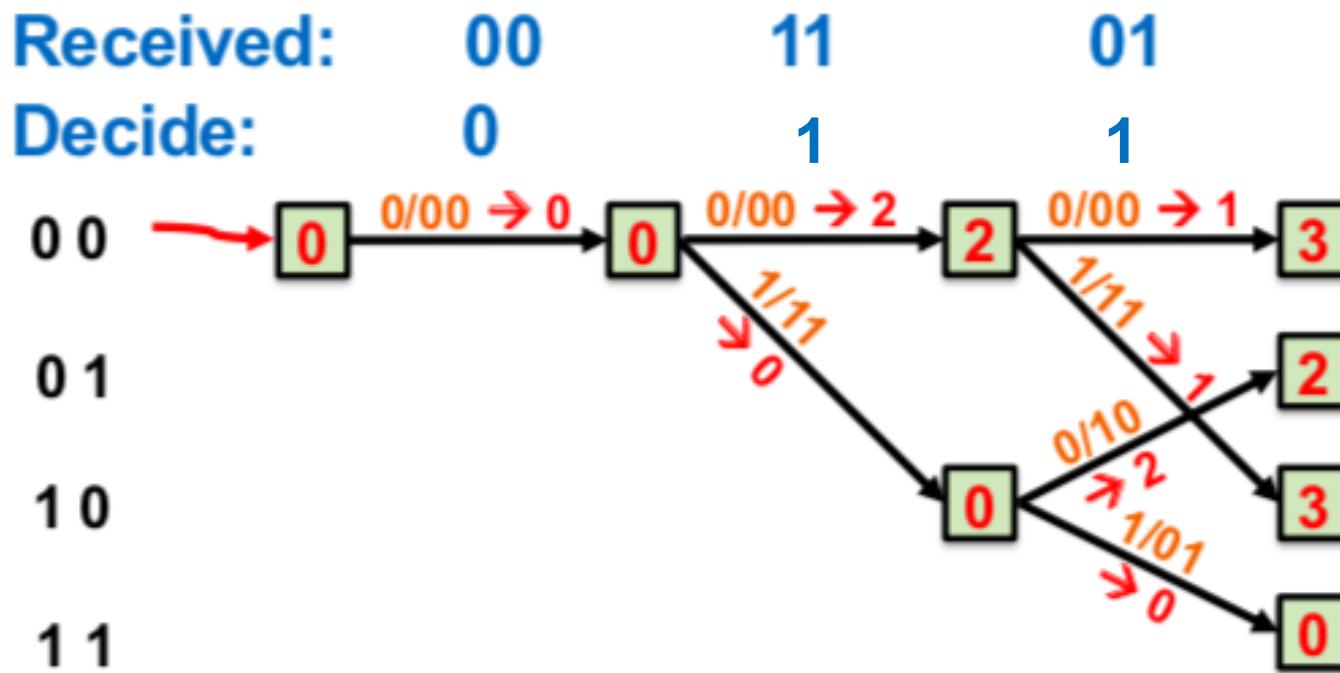
# Hard-decision: survivor path

- The paths that are retained by the algorithm are called survivor
- Some branches are not the part of any survivor: remove them



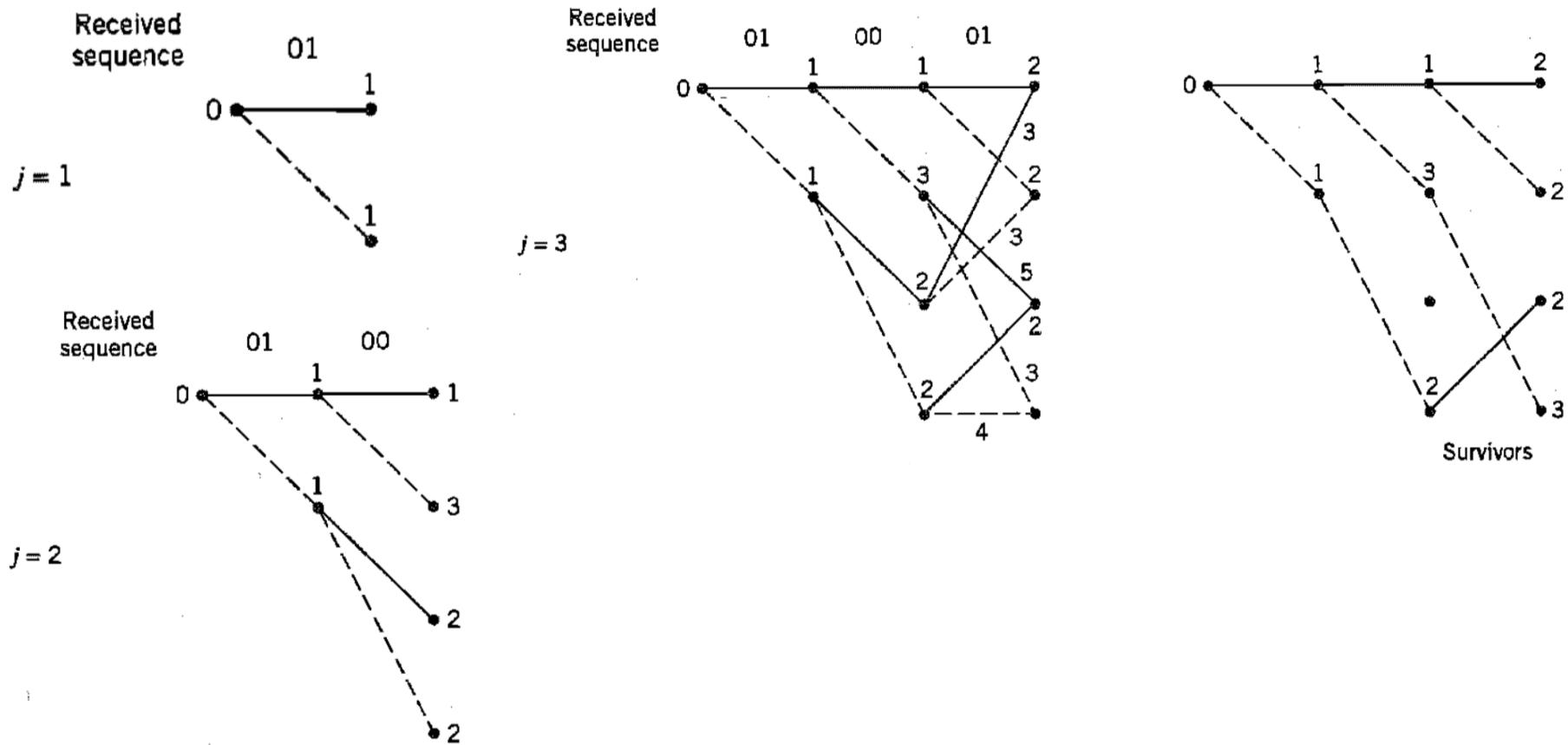
# Hard-decision: estimate $\hat{m}$

- Choose the survivor path with lowest metric
- Estimate  $\hat{m} = 0\ 1\ 1$



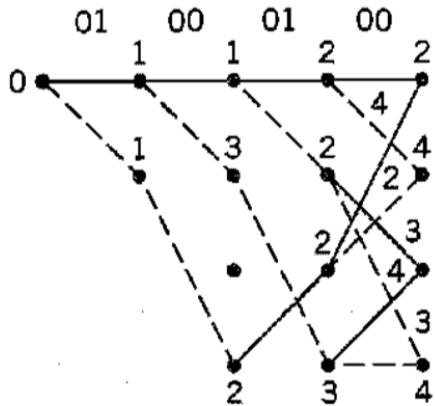
# Hard-decision: Example (1)

- E.g., transmitted codeword:  $c = \{00, 00, 00, 00, 00\}$  and received sequence:  $r = \{01, 00, 01, 00, 00\}$

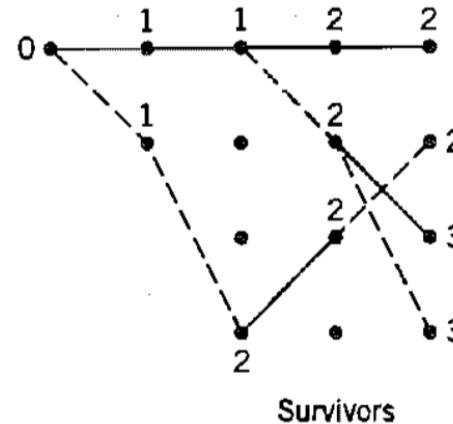


# Hard-decision: Example (2)

Received sequence



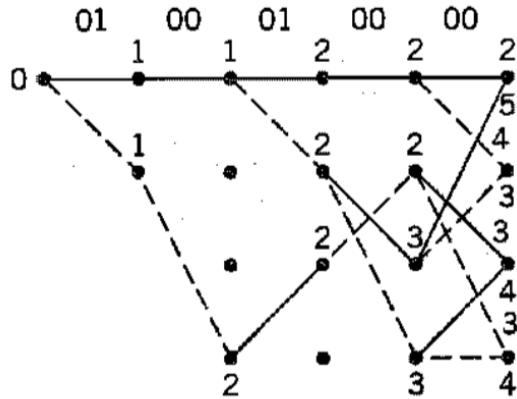
$j = 4$



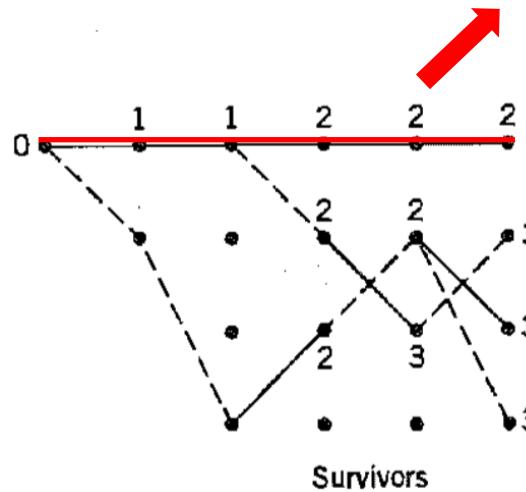
Survivors

Choose the path with  
lowest metric

Received sequence



$j = 5$

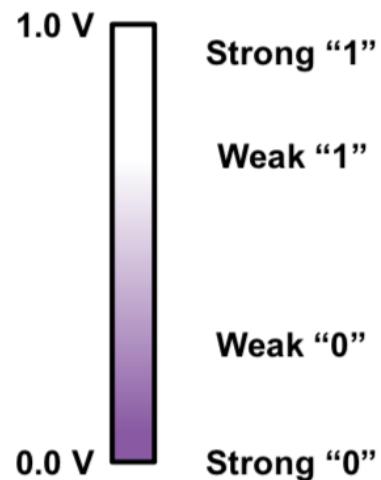


Survivors

$\hat{c} = \{00, 00, 00, 00, 00\}$

# Viterbi Decoding: Soft-decision (1)

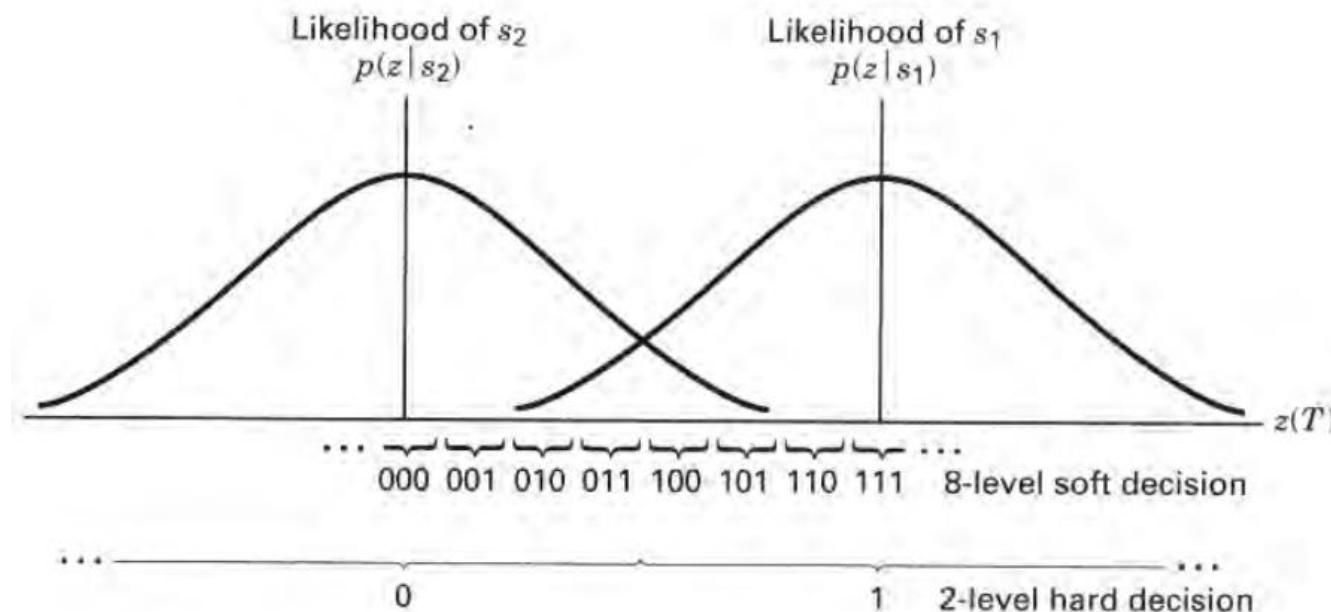
- Coded bits are actually continuously-valued “voltage” between 0V and 1 V



- Hard-decision decoding digitize each voltage to “0” and “1” by comparison against threshold voltage
  - Lose information about how “good” the bit is
    - Strong “1” (0.99V) treated equally to weak “1” (0.51V) with threshold of 0.5V

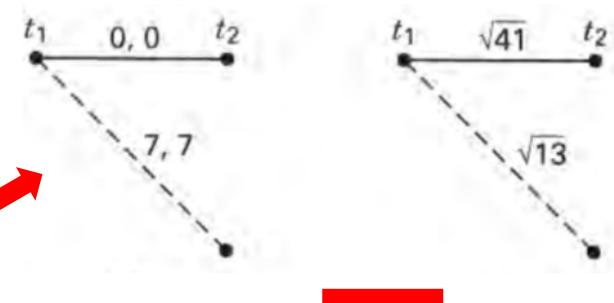
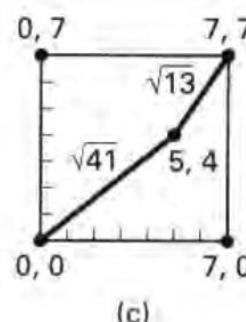
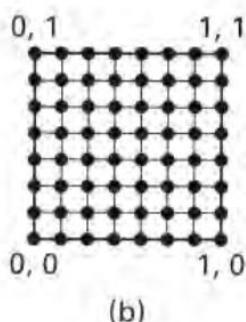
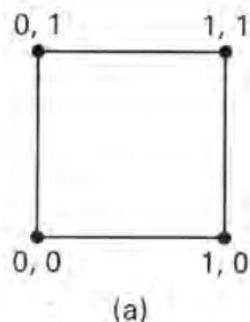
# Viterbi Decoding: Soft-decision (2)

- Soft-decision requires a stream of “soft bits” where we get not only the 1 or 0 decision but also an indication of how certain we are
  - E.g. 000 (definitely 0); 001 (probably 0); 010 (maybe 0); 011 (guess 0); 100 (guess 1); 101 (maybe 1); 110 (probably 1); 111 (definitely 1)
  - We call the last two bits are the “confidence” bits



# Viterbi Decoding: Soft-decision (3)

- For a rate 1/2, the demodulator delivers two code symbols at a time to the decoder
  - For hard-decision (2-level), each pair of received codes can be depicted on a plane (Fig. a)
  - For 8-level soft decision, each pair of symbols can be represented on an spaced 8 level by 8 level plane (Fig. b)
- Soft-decision branch metric: using Euclidean distance (Hamming distance metric cannot use because of its limited resolution)
- E.g. fig. c, a pair of noisy code-symbol values is the point (5,4). What's Euclidean distance?



- Same path metric computation
- Same Viterbi algorithm

# Error Correcting Capability (1)

- How many bit errors can be corrected?



- Using the free distance  $d_{\text{free}}$  to calculate the error-correcting capability of the code

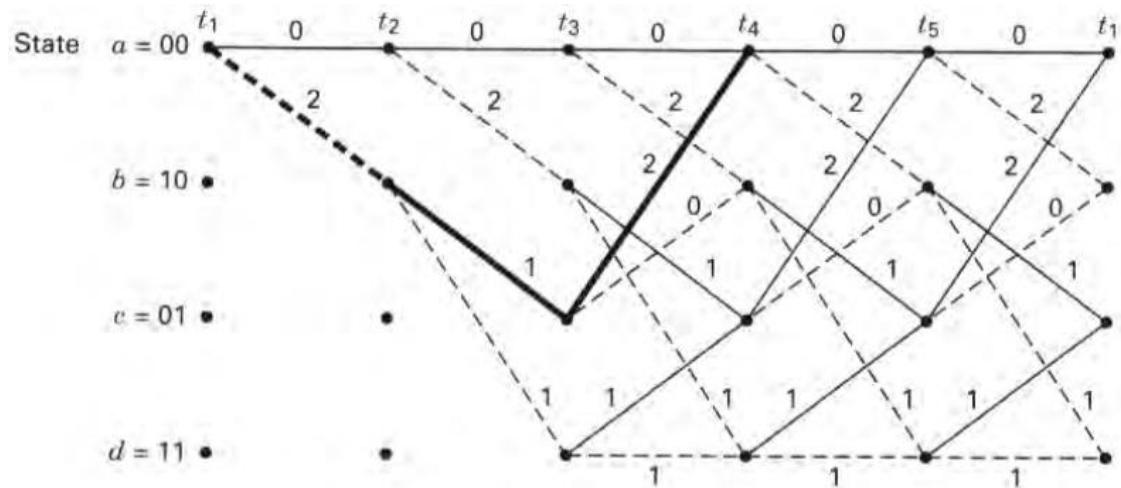
- Free distance = minimum Hamming distance between each of possible codeword sequences and all-zeros sequence
- A convolutional code with  $d_{\text{free}}$  can correct  $t$  errors if and only if  $d_{\text{free}}$  is greater than  $2t$ .

- E.g., Trellis with  $K = 3$

Find the path with smallest non-zeros path metric



$d_{\text{free}} = 5$ : we can correct 2 errors



# Error Correcting Capability (2)

- The value of  $d_{\text{free}}$  depends on the constraint length  $K$ .

Constraint length ( $K$ )	Free distance ( $d_{\text{free}}$ )
2	3
3	5
4	6
5	7
6	8
7	10

**Source:** A. J. Viterbi and J. K. Omura, Principles of Digital Communication and Coding, McGraw-Hill Book Company, New York, 1979, p. 251

# Performance of CC

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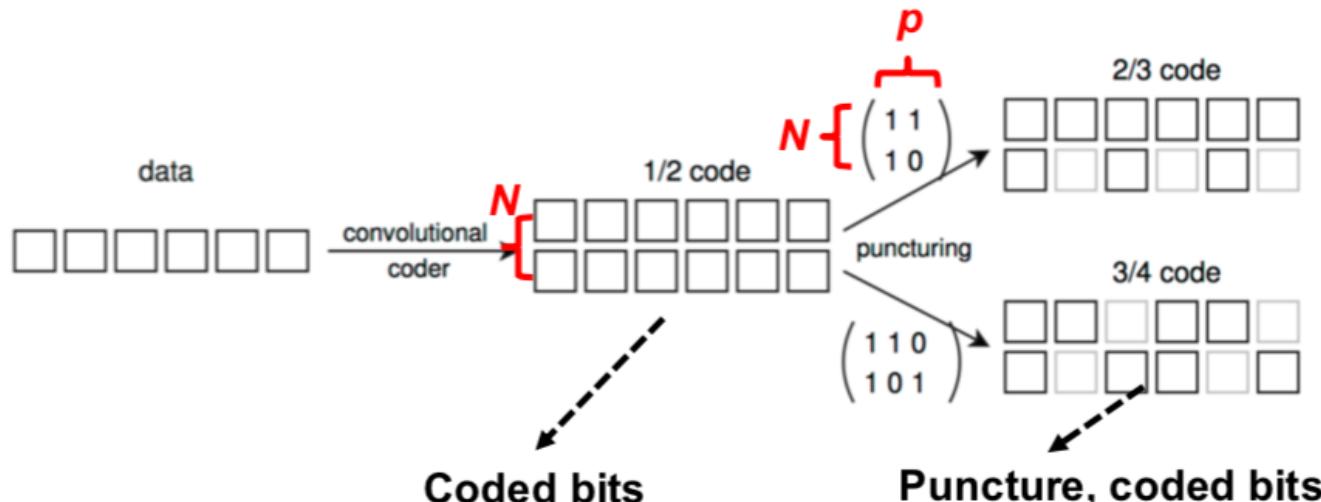
- Performance of CC depends on the coding rate and the constraint length
  - Longer constraint length  $K$ 
    - More powerful code
    - More coding gain
    - More complex decoder
    - More decoding delay
  - Smaller coding rate  $R_c = k/n$ 
    - More powerful code due to extra redundancy
    - Less bandwidth efficiency

# Changing code rate: puncturing

- How to change coding rate?
- E.g. we have a coding rate  $R_c = 1/2$ ; how to change it into a higher coding rate of  $2/3$ . There are two ways
  - Reconstruct the encoder by using an input and output multiplexer: hardware
  - Use **puncturing technique**: software → more convenient
- Idea: delete some bits in the original low-rate coded bits
- Decoding: same Viterbi algorithm (decrease error correction capability)

# Punctured Convolutional Code

- Using puncturing table (a  $N \times p$  matrix) to indicate which bits to include
  - Contains  $p$  columns and  $N$  rows;  $p$  is puncturing period
  - If 1, the corresponding code bit is a part of punctured code
  - If 0, delete the corresponding code bit
- The total number of 1's in the matrix is  $p + L$ ; with  $L = 1, 2, \dots, (N-1)L$
- For  $p$  input information bits, there are  $p + L$  output coded bits. Thus, the rate of the punctured convolutional code is  $p/(p + L)$



# Punctured Convolutional Code: Example (1)

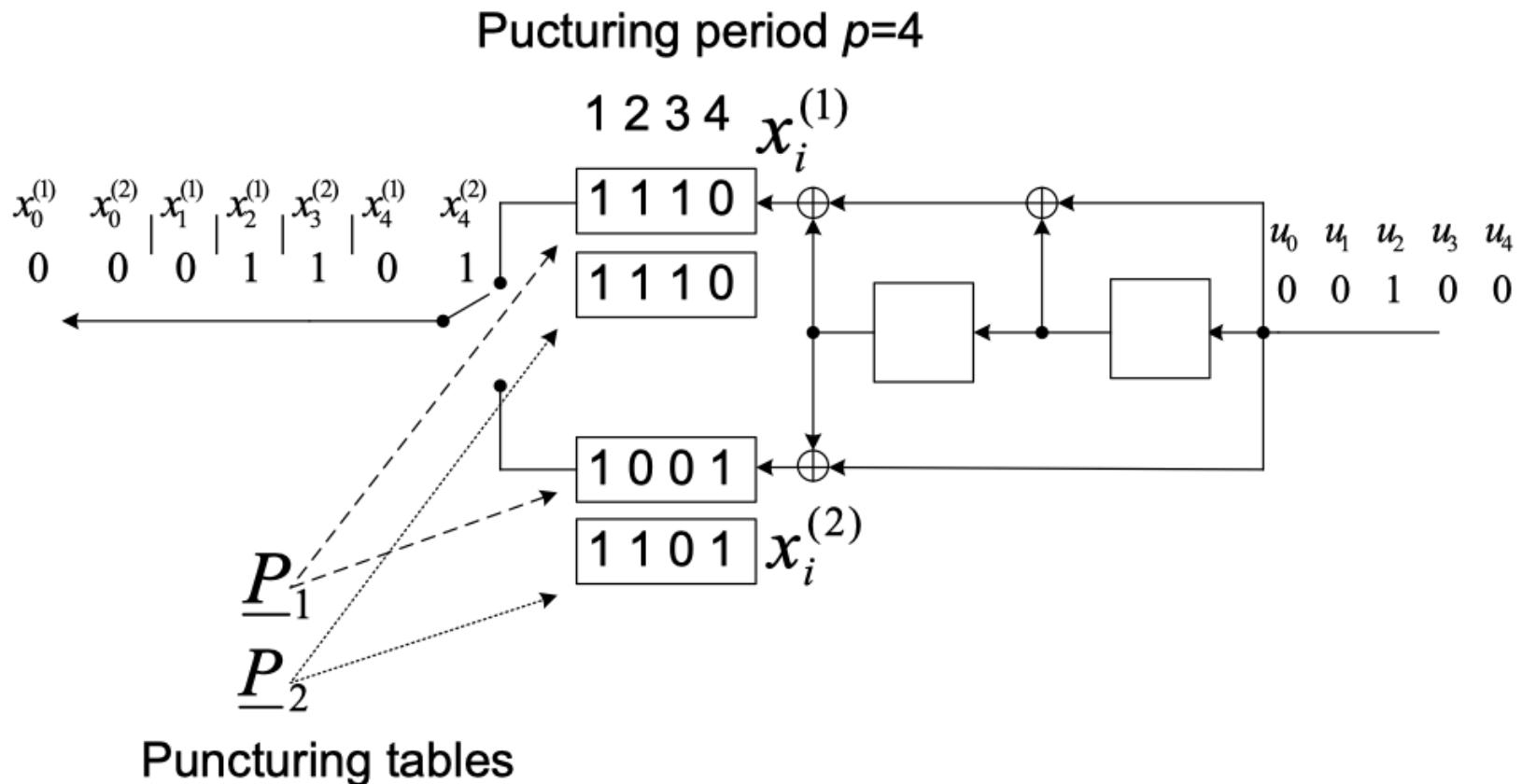
- E.g., message bits  $m = \{0, 0, 1, 0, 0\}$ , coded bits  $c = \{00, 00, 11, 01, 11\}$ , coding rate  $R_0 = 1/2$
- $c$  is punctured using two different puncturing tables (matrices) with the puncturing period is  $p = 4$

$$\underline{P}_1 = \begin{pmatrix} 1 & 1 & 1 & 0 \\ 1 & 0 & 0 & 1 \end{pmatrix}, \quad \underline{P}_2 = \begin{pmatrix} 1 & 1 & 1 & 0 \\ 1 & 1 & 0 & 1 \end{pmatrix}$$

- Using  $\underline{P}_1$ , 3 out of 4 code bits of the mother code are used, the others are discarded, i.e.,  $c = \{00, 0x, 1x, x1, 11\} = \{00, 0, 1, 1, 11\}$ 
  - Coding rate,  $R = 4/5$
- Using  $\underline{P}_2$ , 2 out of 4 code bits of the mother code are used, the others are discarded, i.e.,  $c = \{00, 00, 1x, x1, 11\} = \{00, 00, 1, 1, 11\}$ 
  - Coding rate,  $R = 4/6 = 2/3$

# Punctured Convolutional Code: Example (2)

- Encoder of a rate 1/2 code is punctured to a rate 4/5 (top puncturing table) or a rate 2/3 code (bottom puncturing table)

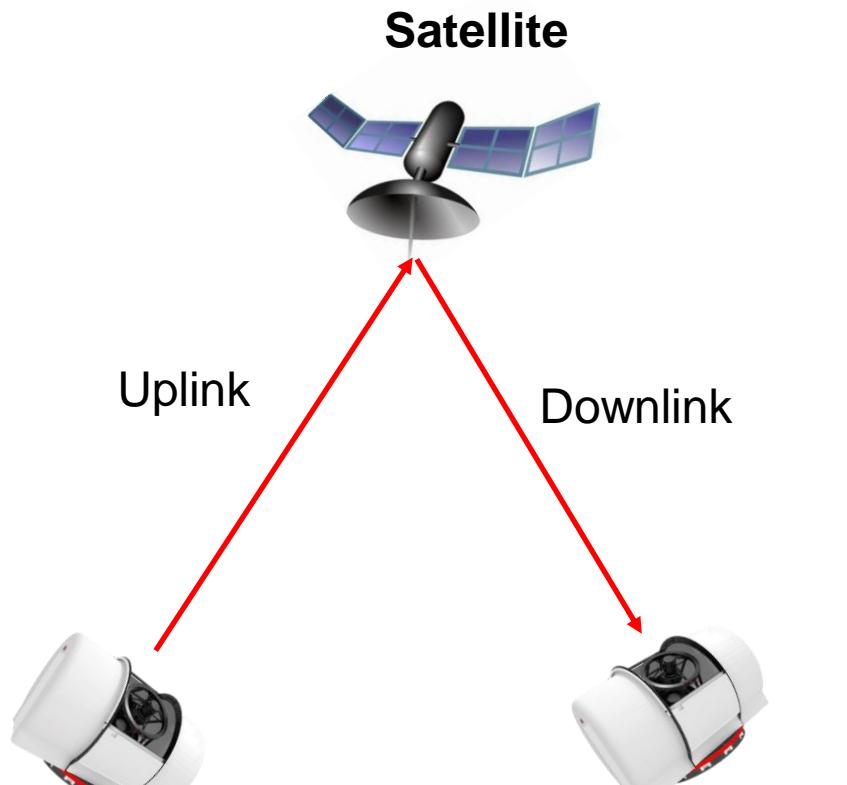


# Rate-compatible punctured CC

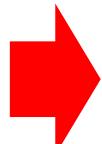
- How to design convolutional code in adaptive systems (with variable-rate coding)?
  - Puncturing technique is used for change code rate
  - Using rate-compatible restriction: all code bits of higher rate punctured code of the family (from a mother code) are used by the lower rate codes
- This way guarantees smooth transition between different code rates in the systems using adaptive FEC codes
- Rate-compatible punctured convolutional (RCPC) codes
  - if higher rate codes are not sufficiently powerful to decode channel errors, only supplemental bits which were previously punctured have to be transmitted in order to upgrade the code



# Part 2: FSO-based Satellite Systems



- Satellite systems widely used in:
  - Navigation
  - Broadcasting
  - Disaster recovery
- Classification: LEO (between 160 – 2000 km), MEO (between LEO and GEO), and GEO (~36,000 km)
- FSO-based satellite to provide high-speed connections (~Gbps)
- Challenge considered in my research: atmospheric turbulence (its effect is up to 40 km above the sea level)



Proper error control methods are needed

# Solutions (1): Error Control Methods

- In FSO-domain, there are two popular error control methods: ARQ protocols and FEC codes
- ARQ: retransmission
  - When the channel error rate is high: not efficient due to the increased frequency of retransmissions.
  - In satellite systems: delay is the important issue due to retransmissions.
    - Terrestrial (2 km)  $\sim 6.67 \mu\text{s}$ , LEO satellite(2000 km)  $\sim 6.67 \text{ ms}$
- FEC code: add redundancy to correct errors
  - When the channel less noisy: decrease the throughput due to adding redundancy
  - If the errors are uncorrectable by FEC: lose the reliability

# Solutions (2): Hybrid FEC-ARQ (HARQ)

- HARQ: hybrid between FEC and ARQ
- FEC: try to correct the errors first in order to reduce the frequency of retransmissions.
- ARQ: is used for retransmission if the errors are uncorrectable by FEC.



- In this way, it is possible to achieve a higher reliability than a FEC alone and lower delay than ARQ alone

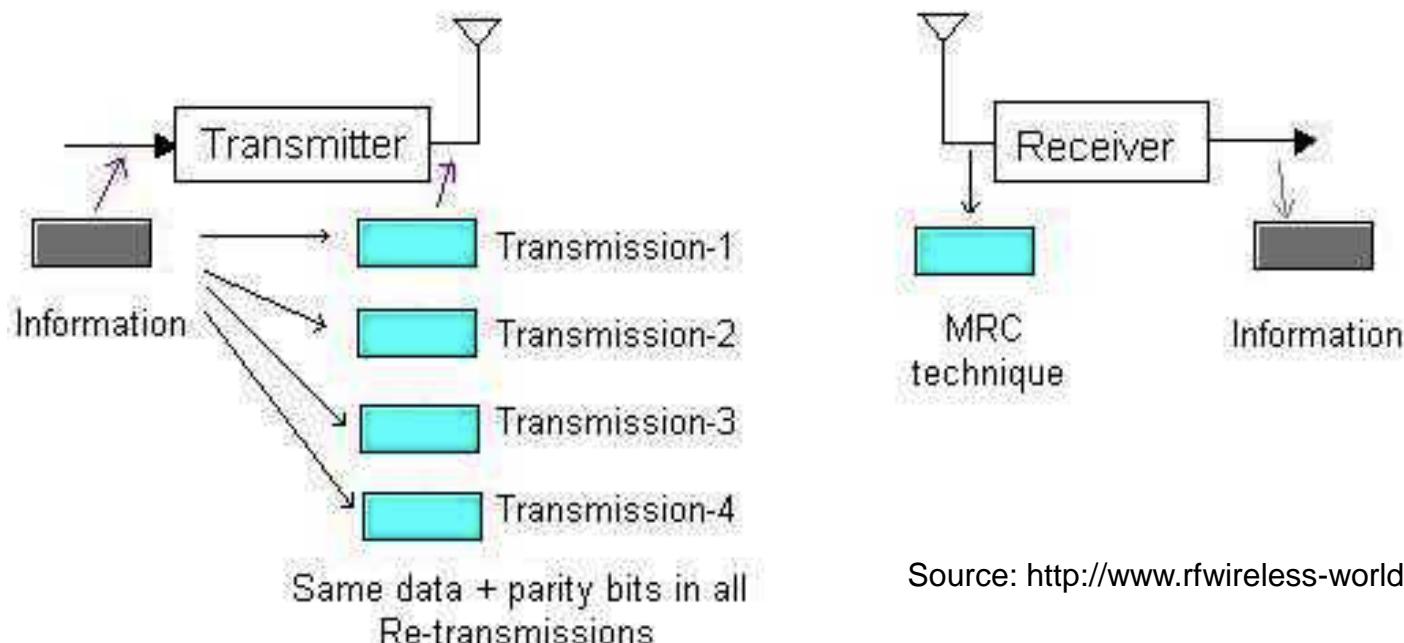
# Review of HARQ Protocol

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- Type I-HARQ: always discards corrupted frames while they still contains some useful information → **not efficient**
- Type II-HARQ: is an advanced form of HARQ which uses the concept of frame combining
- Frame combining: the corrupted frames will be stored in the receiver's buffer to be combined with other retransmissions to enhance the correction performance
- Type II-HARQ can be classified into 2 types: chase combining (CC) and incremental redundancy (IR)

# Chase Combining

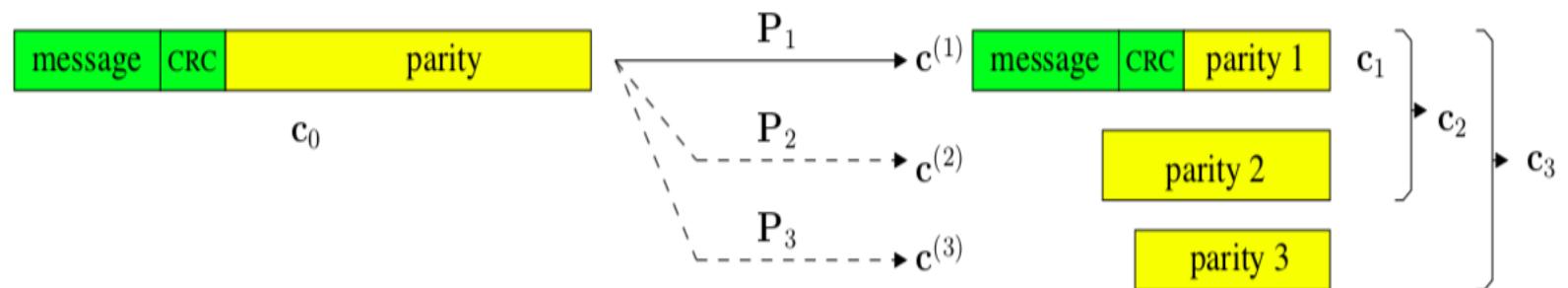
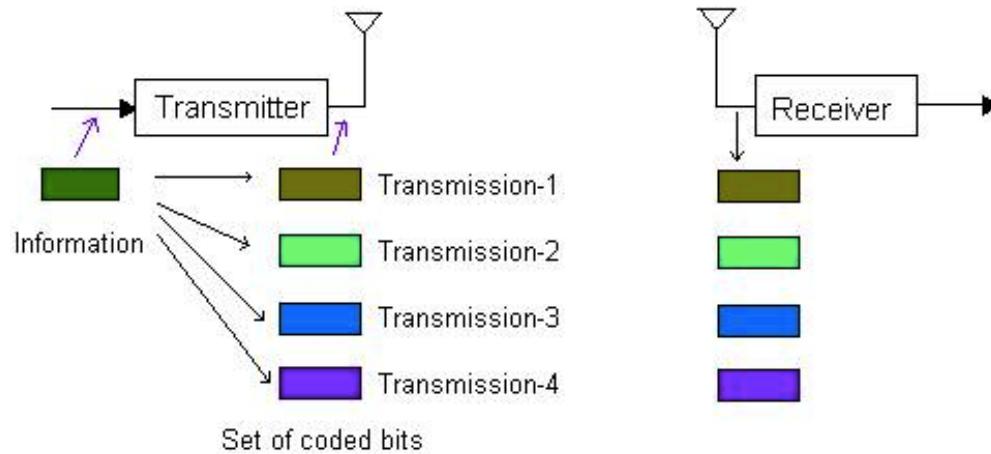
- CC-HARQ: same frames including FEC code are retransmitted each time and retransmitted frames will be combined to obtain the correctable information thanks to Maximum-ratio combining (MRC) technique



Source: <http://www.rfwireless-world.com>

# Incremental Redundancy (IR)

- IR-HARQ: effective code rate is gradually lowered until received frame is decoded correctly → **more efficient**



# Literature Survey of HARQ

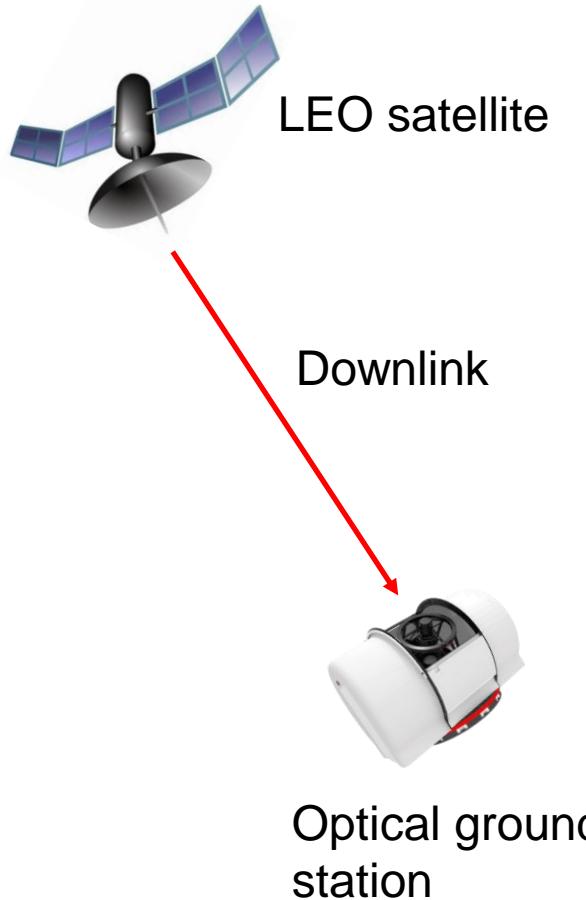
- In FSO domain (IEEE-Journal)

Reference	Main contribution	Type of FEC	Type of ARQ
[1] - 2011	Type II (CC) HARQ over Log-normal	Block code	Stop-and-wait
[2] - 2012	Type I and II HARQ over Gamma-gamma	Block code	Stop-and-wait
[3] - 2014	Type I and II HARQ in FSO with pointing error	Block code	Stop-and-wait
[4] - 2016	Type II HARQ in RF-FSO	Block code	Stop-and-wait
[5] - 2017	Type II RF-FSO multi-hop with HARQ	Block code	Stop-and-wait

- Problems:

- Most of them considered the employ of HARQ in PHY layer point of view.
- Hardware implementation in high-speed connection (~Gigabit) at PHY: big challenge, showed in [6] → should be employed at the link layer (faster).
- The stop-and-wait ARQ in FSO: not efficient, demonstrated in my previous works → should be replaced by sliding window ARQ
- AMC should be employed to improve system performance

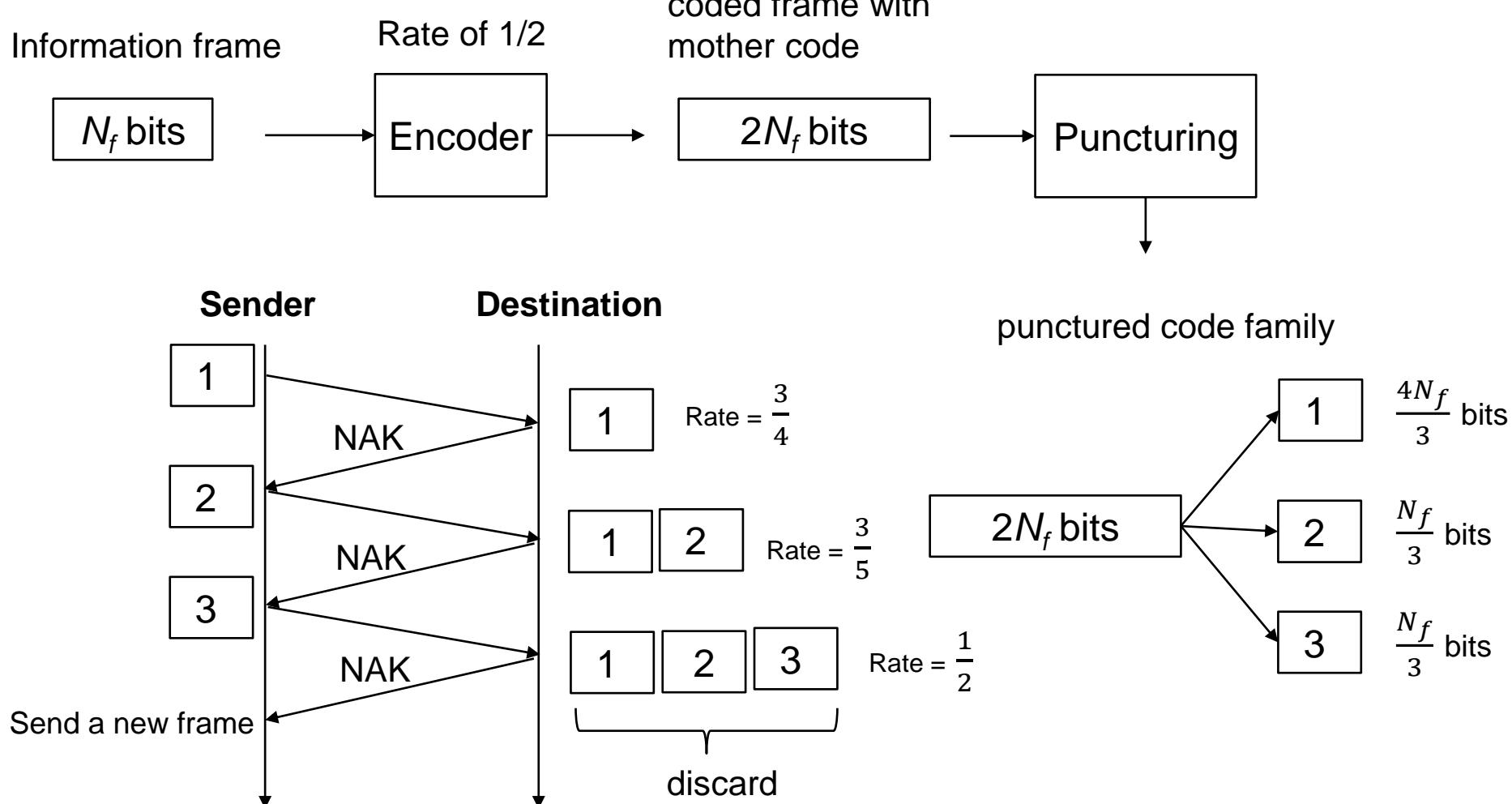
# My study



- **Link layer: truncated IR-HARQ**
  - FEC: rate-compatible punctured convolutional codes
  - ARQ: sliding window ARQ
- **PHY layer:**
  - Adaptive Modulation and Coding (AMC)
  - Burst transmission with adaptive number of frames
  - Channel model (source: NICT experiment): Gamma-gamma is the best fitting model for downlink in LEO satellite systems

# How it works? (1)

- LL: for each frame



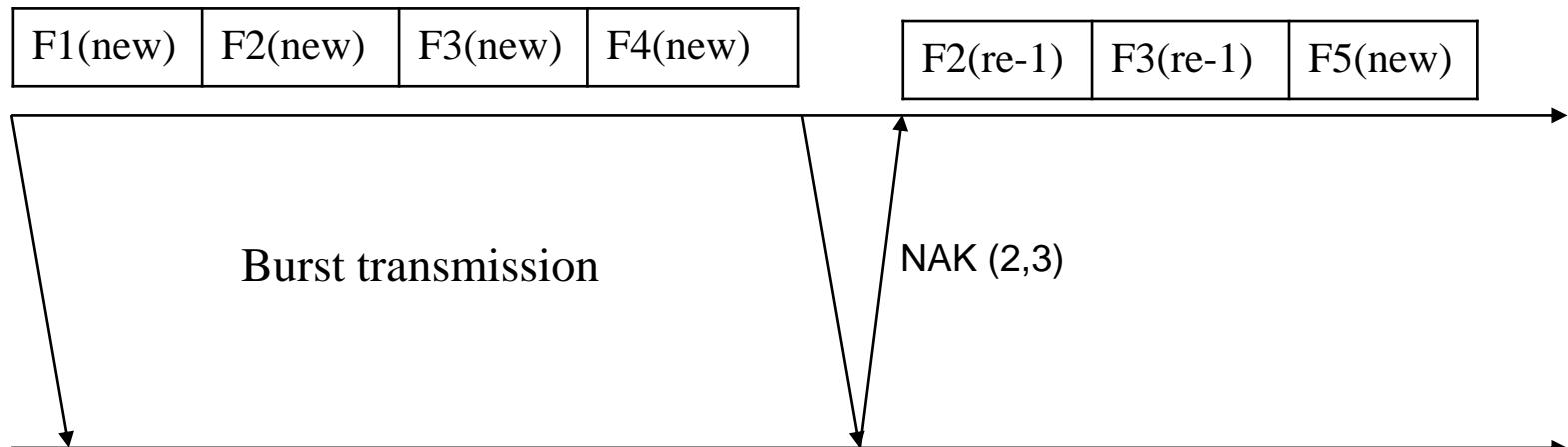
# How it works? (2)

- PHY: Burst transmission

Equal-size frames?



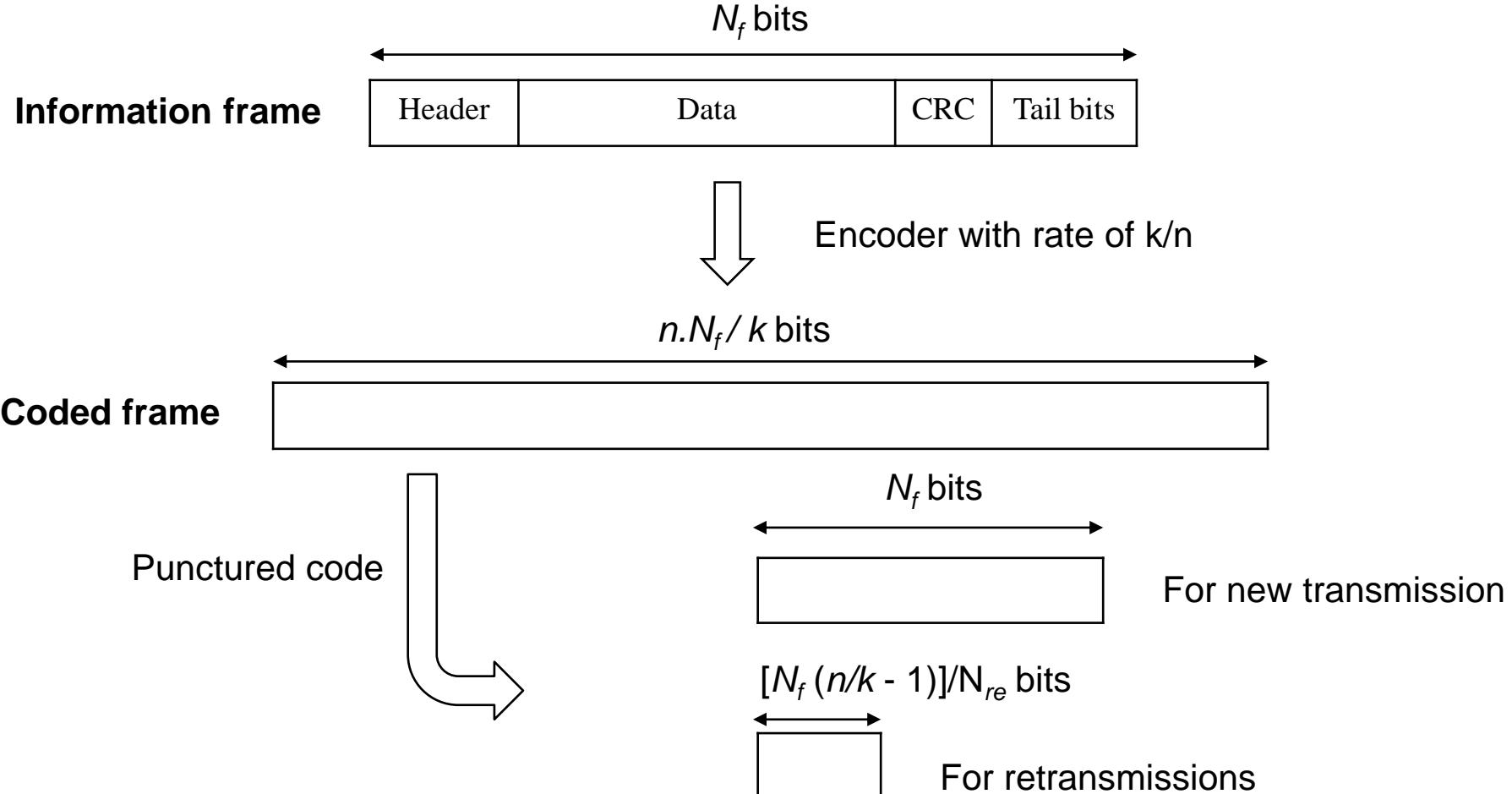
Transmitter



Receiver

# Frame Design

- Frame structure



Maximum no. of retransmissions:  $N_{re}$

# Burst Transmission Design

- How to design a burst transmission?

Header	Payload (adaptive number of frames)
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Mode 1: BPSK	1 <sup>st</sup>	1 frame					
Mode 2: QPSK	1 <sup>st</sup>	2 <sup>nd</sup>	2 frames				
Mode 3: 8-QAM	1 <sup>st</sup>	2 <sup>nd</sup>	3 <sup>rd</sup>	3 frames			
Mode m: 2 <sup>m</sup> -QAM	1 <sup>st</sup>	2 <sup>nd</sup>	3 <sup>rd</sup>	...	m-1 <sup>th</sup>	m <sup>th</sup>	m frames

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## Others

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**Thank you for your attention!**  
**(Q&A)**

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