



ALBERT-LUDWIGS-
UNIVERSITÄT FREIBURG

Communication Systems

ISDN

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Plan

- ▶ Welcome to a completely different world
- ▶ Telephony protocols are defined through standards
- ▶ Standards in telecommunication
- ▶ In telephony world mostly not talked on “protocols” but interfaces
- ▶ Digital telephony networks – from analogous source to digitized data streams
- ▶ Basic overview on ISDN – Integrated Services Digital Network today, switching to mobile telephony next lecture

Standards in Telecommunication

- ▶ Interfaces are well-defined connection points where different parts of the infrastructure/equipment talk to each other in a certain way
- ▶ International standardization body is ITU (International Telecommunication Union www.itu.int)
- ▶ Process of standardization completely different to the workflows in Internet bodies
 - no bottom up, but top down decisions
 - exclusive club of the big (state monopoly) Telcos
 - high annual fees
 - much less information publically available then for IP and other open protocols

Standards in Telecommunication

- ▶ Because of the old (nation state) monopolies there are many differences within the several networks
 - Numbering schemes
 - Acoustical indication of dial states (busy, line-free, ...)
 - Different use, assignment of the (wireless) frequency spectrum
 - Not really compatible equipment (branch exchanges, ...)
 - every firm tries to use their own subset of “standards”
- ▶ With the introduction of digital networks (ISDN and mobile) agreement on global standards started

Standards in Telecommunication

- ▶ Inter connecting of voice streams has lots of technical problems
- ▶ Up to 1980s computerized switching centers but analogous voice connections
 - fault-prone to jamming and noise
 - regeneration means amplification of noise too
- ▶ Allow data connections over telephony networks
- ▶ Next step: Fully computerized switching centers
 - out of band signaling of call setup
 - digital voice streams allow better/perfect regeneration

ISDN – Integrated Services Digital Network

- ▶ The “future” of digital wide area networks in the 80th until mid 90th
- ▶ The development of digital switching networks led to standardization and integration of additional services into the same network
 - three virtual multiplex channels over the same two wire infrastructure (if standard Basic Rate Interface BRI used)
 - digital telephony (two independent lines on basic rate interface)
 - fax, telex
 - video telephony (H.323 devices may use ISDN as transport layer for their applications)
 - data communication of 64 or 128kbit/s

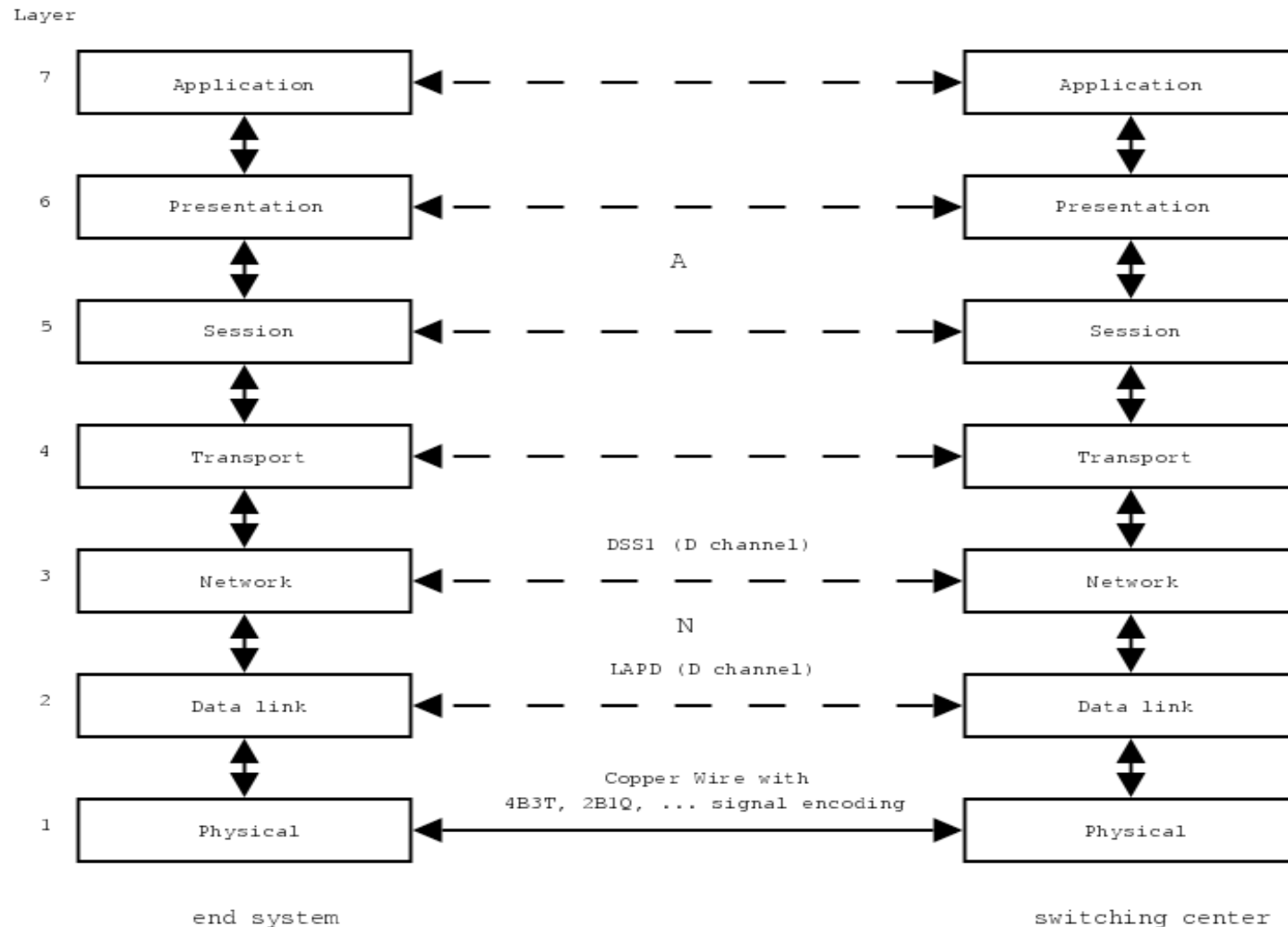
ISDN – Integrated Services Digital Network

- ▶ Prerequisite for ISDN was digitalized infrastructure
- ▶ The ISDN standard was defined in the early 1980s by the ITU
 - several national standards evolved, 1TR6 in Germany, NI-1/2 in United States, DACS in UK, ...
 - DSS1 is the “EURO-ISDN” used in many other countries too available from 1993
 - EURO ISDN was defined by the new founded ETSI (European Telecommunication Standards Institute in 1988)

ISDN – Integrated Services Digital Network

- ▶ ISDN is commonly used in all European countries since 2000
 - all switching centers use ISDN backends
 - so called “analogous” telephony devices (POTS – plain old telephony service) are converted to digital service at the local switching center
 - 50% of the European BRI connections are in Germany
 - Germany has a 30% worldwide share

ISDN – and the OSI protocol stack (mostly D channel)



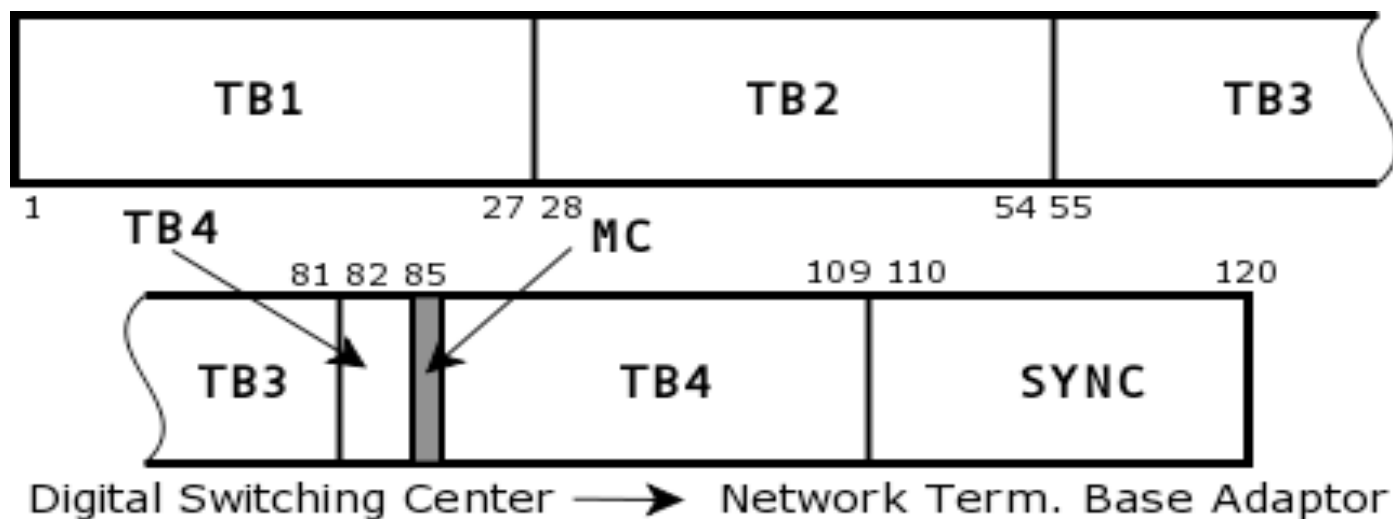
ISDN – Basic Rate Interface

- ▶ Example for physical layer
- ▶ Alternate encoding: 2B1Q – 2 bit digital to 1 baud quaternary representation
- ▶ 2B1Q transmission can be simply described as an amplitude modulation scheme for DC pulses
- ▶ Ordering of data blocks depends on the encoding used

Bits	Voltage
00	-2.50
01	-0.83
10	2.50
11	0.83

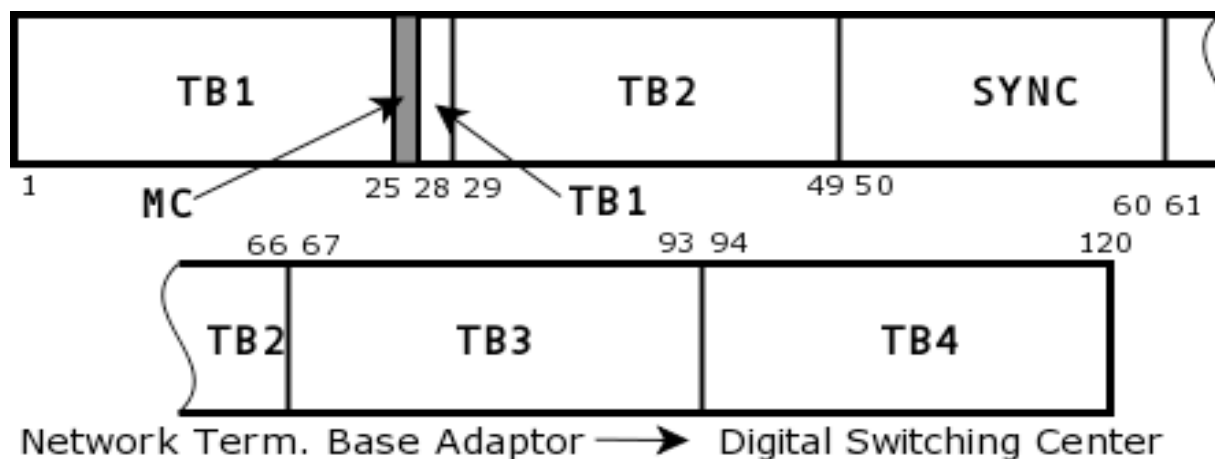
Uk0 – bit streams from switching center to NTBA

- ▶ Data Link Layer in ISDN
- ▶ Each frame consists of 120 ternary steps
 - $2*B+1*D$ takes 108 steps in 4 ternary blocks (tb) with 27 steps each
 - sync channel occupies 11 steps and a “maintenance” channel (mc) 1 step



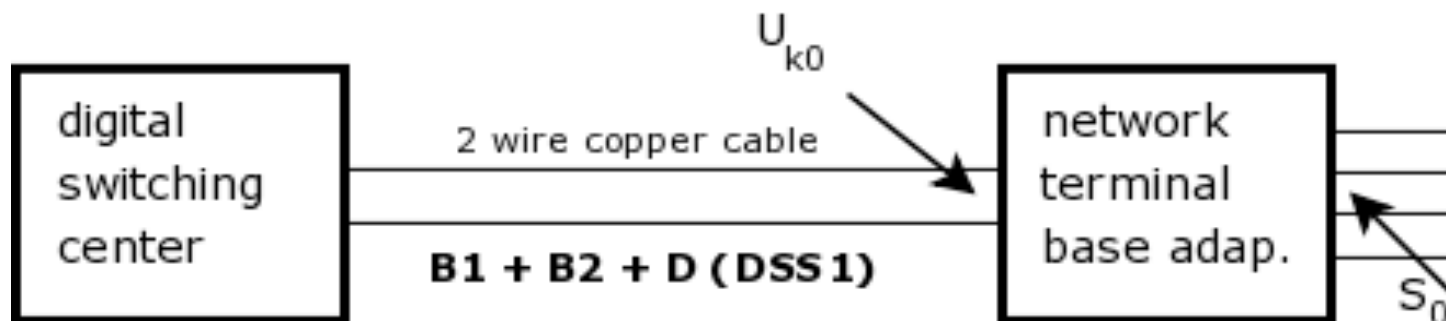
Uk0 – bit streams from NTBA to switching center

- ▶ Connection is full-duplex over the two wires
 - echo compensation and terminating set is needed
 - NTBA splits the data streams to separate up and down onto the S0 bus



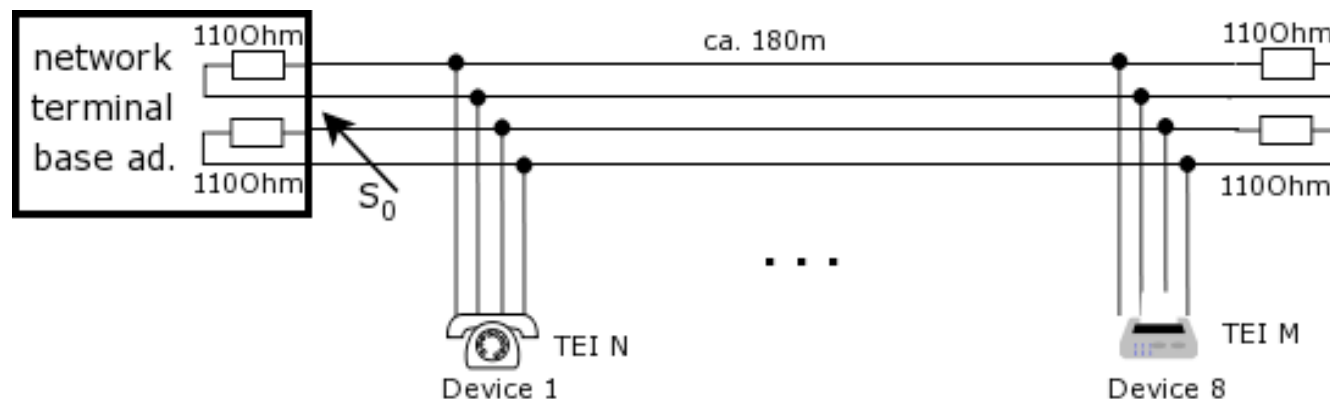
ISDN – Basic Rate Interface

- ▶ Instead of the traditional wall socket a NTBA (network terminal base adapter) is needed at end users site
- ▶ NTBA provides the S0 bus to which end user devices are connected
 - unidirectional – on pair of wires for each direction
 - allows up to 12 wall sockets, 8 ISDN devices (or analogous devices via a/b converter)
 - provides device power up to 4,5W



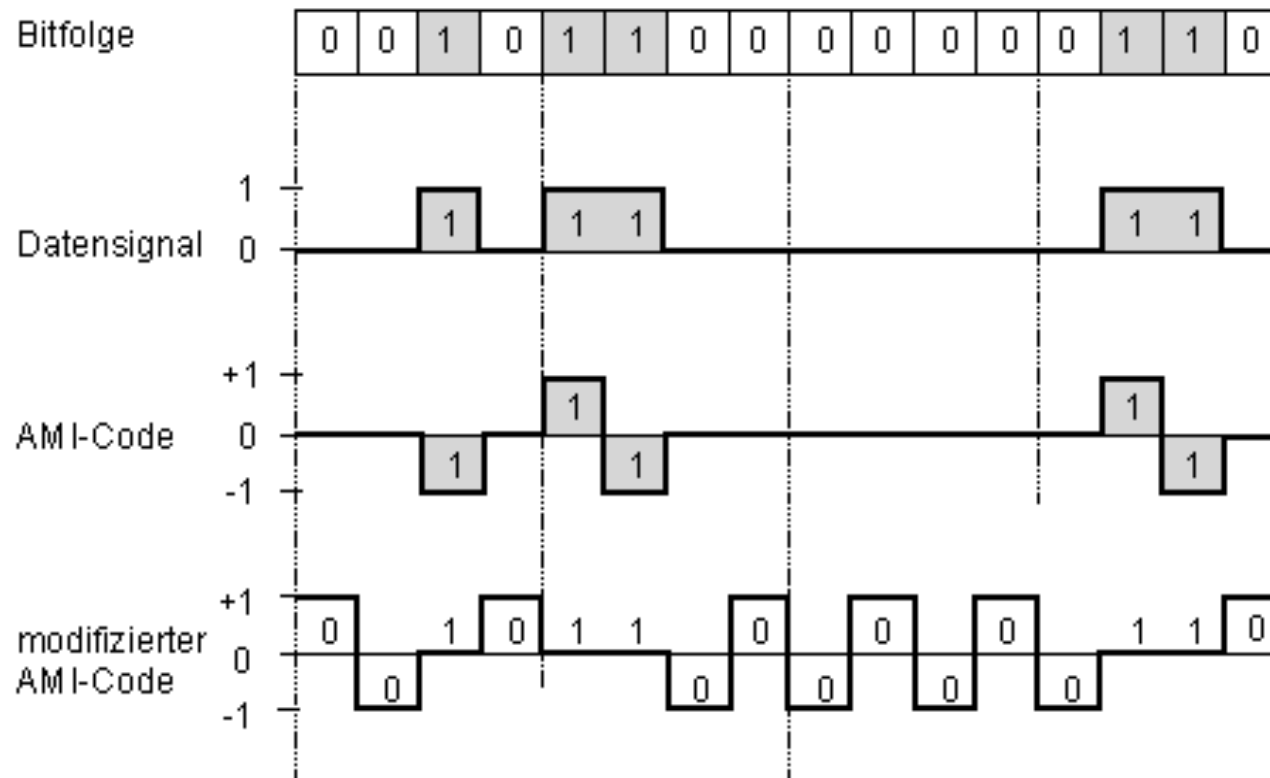
ISDN – S0

- ▶ Provides the same B and D channels as Uk0
 - maintains the step and octet frequency
 - handles the device plugging and device activation, deactivation
 - has to be terminated with resistors of 110 Ohm
 - uses modified AMI code with currents of -0,75 and 0,75V



S0 – AMI code

- ▶ Modified AMI code (avoid long sequences of symbols of the same type)

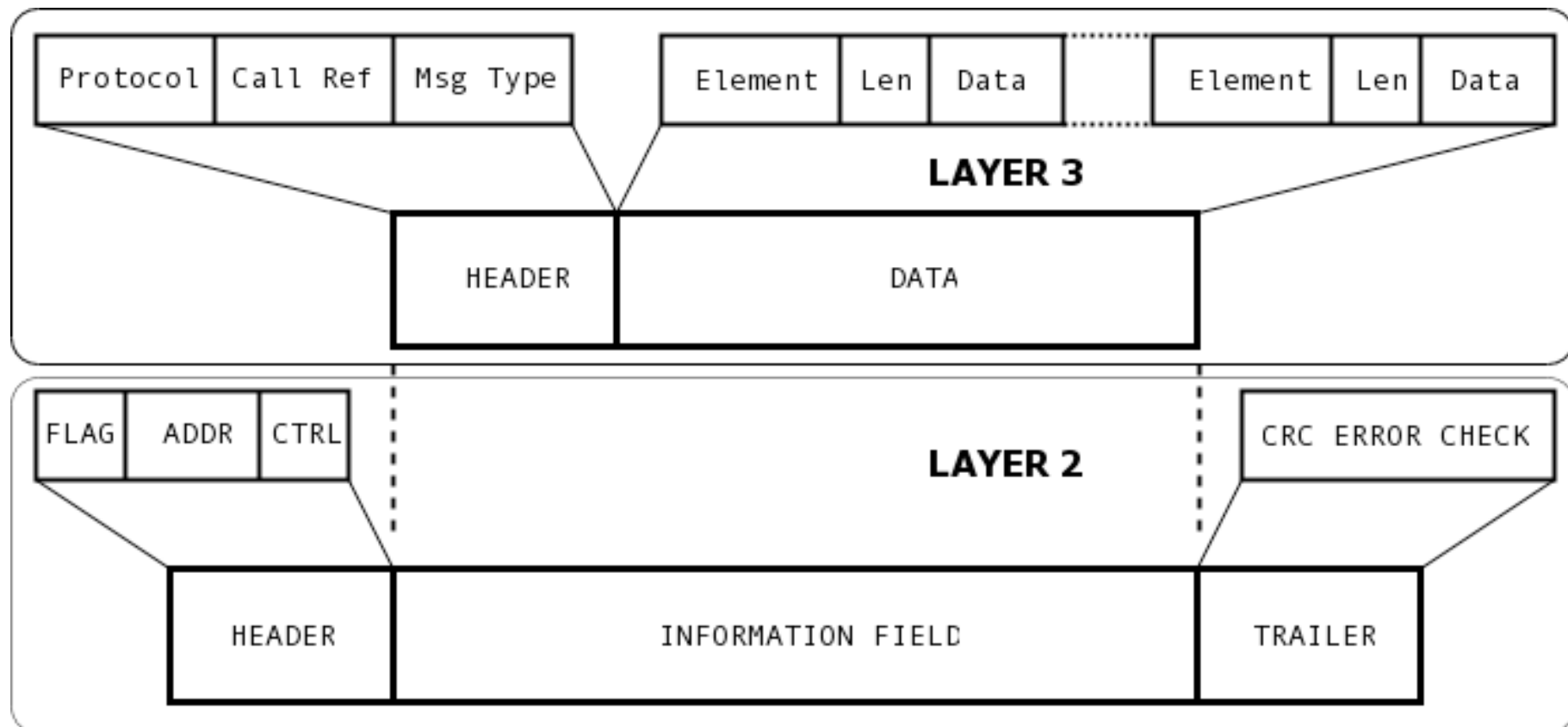


Data Link Layer for the D channel

- ▶ No distinct layering for B channels – PCM or data directly put into frames as shown on previous slides
- ▶ LAPD – Link Access Procedure on D channel
 - derived from High-Level Data Link Control Protokoll (HDLC)
 - broadcasts only for network termination device
- ▶ D2 frame margin – octet of binary pattern: 01111110
- ▶ Keeping of frame sequence
- ▶ Error discovery
- ▶ Multiplexing of more than one logical D2 connections
- ▶ Flow control

Higher Layer Protocols for the D channel

- ▶ ITU Recommendation Q.921



Layer 2 for the D Channel

- ▶ Flag
 - character is part of the Header information, hexadecimal 7E
- ▶ Address is two bytes (octets) long, and consists of three fields
 - Service Access Point Identifier (SAPI)
 - Command/Response (C/R) bit
 - Terminal Endpoint Identifier (TEI)

Layer 2 for the D Channel

- ▶ Control one or two octets (bytes) in length, indicates one of three frame formats
 - information
 - supervisory
 - unnumbered
- ▶ Information carries Layer 3 Call Control (Q.931) data
 - it may carry Unnumbered Information data (TEI assignment) or XID (Connection Management/parameter negotiation) information

Data Link Layer for the D Channel

- ▶ Protocol handles the TEI (Terminal Endpoint Identifier) allocation
 - all devices on S0 using the same bus and have to be addressable
 - TEI assignment is started by the connected devices after successful initialization of physical layer synchronization
 - non automatic assignment uses ID0 – 63, automatic 64 – 126
 - there is a special group TEI 127
- ▶ Protocol elements
 - information lowermost bit is set to 0

Data Link Layer for the D Channel

- ▶ Protocol elements
 - Receive Ready - (01)
 - Set Asynchronous Balance Mode Extended - (6F/7F)
 - Unnumbered Information - (03)
 - Disconnect - (43/53)
 - Unnumbered Acknowledgement – (63/73)
- ▶ Flow control uses sequence numbers for sending and receiving
- ▶ 00:E1:04:00:....
- ▶ Octets #4 for sending and #5 for receiving in the information frame

Data Link Layer for the D Channel – Error Detection

- ▶ D channel protocol uses rather sophisticated error detection protocol
- ▶ Generates frame checksums
- ▶ Generator polynomial
- ▶ $g(x) = (x + 1)(x^{15} + x^{14} + x^{13} + x^{12} + x^4 + x^2 + x + 1)$
- ▶ $g(x) = x^{16} + x^{12} + x^5 + 1$
- ▶ 16 bit frame checksum
- ▶ Inverted residue of binary division
- ▶ $p_1(x) = x^k (x^{15} + x^{14} + \dots + x^2 + x + 1)$
- ▶ $p_2(x) = x^{16} d(x)$

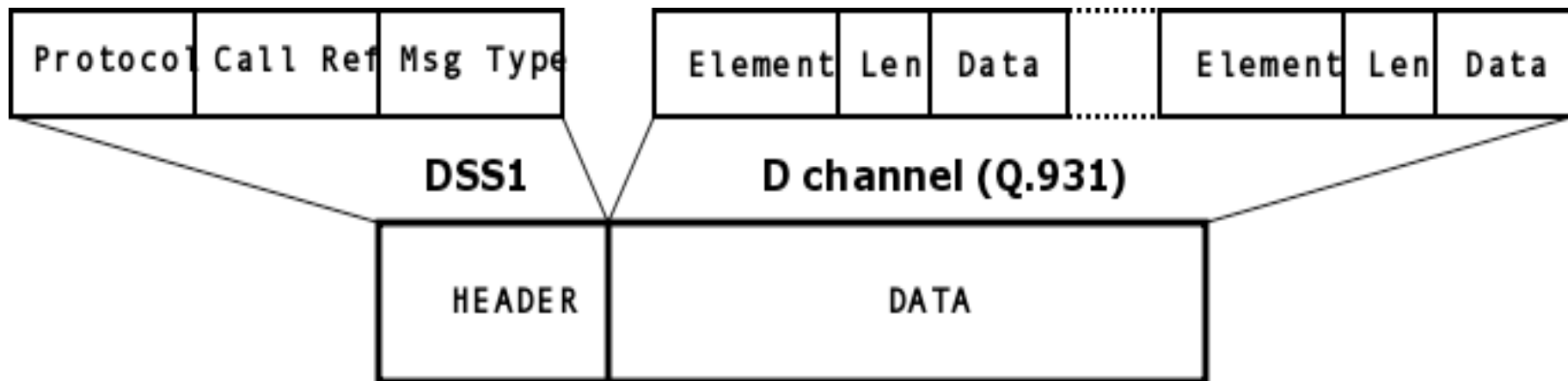
data link layer for the D channel

error detection

- ▶ Checking for added or lost binary zeros
- ▶ Thus cyclic Hamming codes implemented
- ▶ Error detection for one, two and three bit error

network layer for the D channel

- ▶ DSS1 protocol handles the call setup of the calling and called site
- ▶ Call destruction after finishing the session
- ▶ Restarting and parking if required
- ▶ Error handling

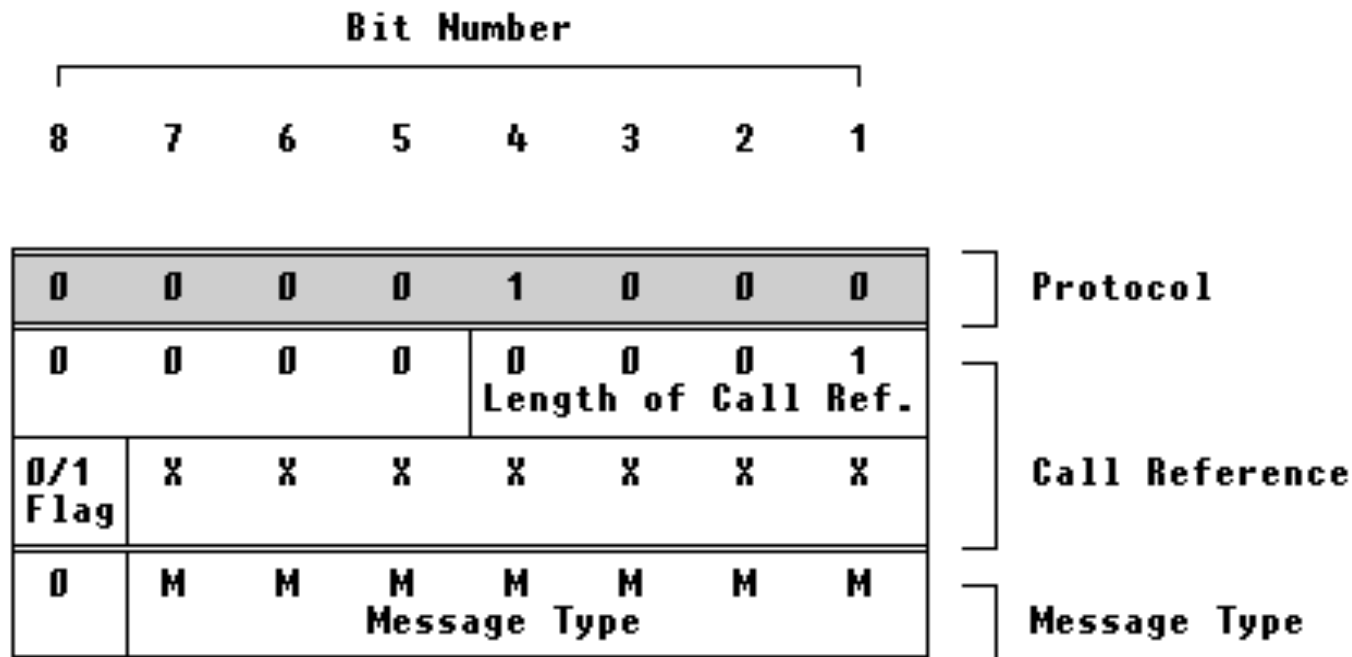


DSS1 layer 3 protocol

- ▶ Protocol Discriminator
 - part of the Layer 3 header information
 - single byte (octet) that is usually set to a value of 00001000 (hexadecimal "08") - meaning Q.931 call maintenance
- ▶ Reference Value consists of either two or three bytes (octets)
 - BRI systems have a 7-bit Call Reference value (127 references)
 - no particular end-to-end significance
 - Either end can assign an arbitrary value
 - used to associate messages with a particular channel connection

DSS1 layer 3 protocol

- ▶ Message Type single byte (octet) that indicates what type of message is being sent/received



DSS1 layer 3 protocol – message types

- ▶ Message Type – four categories
 - Call Establishment
 - Call Information
 - Call Clearing
 - Miscellaneous

DSS1 layer 3 protocol – information elements

- ▶ Each type of message has Mandatory and Optional Information Elements, identified with single byte (octet)
 - bearer Capability (identifies transport requirements of the requested B-Channel)
 - cause (identifies reasons for disconnect or incomplete calls)
 - channel Identification (identifies type and number of B-Channel(s) requested)
 - progress Indicator (indicates status of outgoing call)

DSS1 layer 3 protocol – information elements

- Network Specific Facilities (Useful for North American PRI calls - identifies network type, Carrier ID, Carrier Service Type [WATS/SDN/ASDS,etc.])
 - Calling Party Number (caller ID)
 - Calling Party Number sub address
 - Called Party Number (destination number, type of number[unknown], numbering plan)
 - Called Party Number sub address
- ▶ When Information Elements (IE) consist of multiple octets, the following octet describes how many bytes (octets) are in the Information Element

Literature

- ▶ RSVP
 - http://www.cisco.com/univercd/cc/td/doc/cisintwk/ito_doc/rsvp.htm
- ▶ Telephony (ISDN, ...)
 - E. Pehl, Digitale und analoge Datenübertragung
- ▶ QoS
 - Queueing Disciplines for Bandwidth Management: <http://lartc.org/howto/lartc.qdisc.html>



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